Calibration of acoustic instruments
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David A. Demer • Laurent Berger • Matteo Bernasconi • Eckhard Bethke • Kevin Boswell • Dezhang Chu • Reka Domokos • Adam Dunford • Sascha Fässler • Stéphane Gauthier • Lawrence T. Hufnagle • J. Michael Jech • NaigLe Bouffant • Anne Lebourges-Dhaussy • Xavier Lurton • Gavin J. Macaulay • Yannick Perrot • Tim Ryan • Sandra Parker-Stetter • Sarah Stienesen • Thomas Weber • Neal Williamson
Contents

Foreword......................................................................................................................... 4

Terms, symbols, and units ............................................................................................... 5

1 Introduction .................................................................................................................. 10
  1.1 Scope and outline .................................................................................................... 10
  1.2 Echosounder theory ............................................................................................... 11
      1.2.1 Introduction ...................................................................................................... 12
      1.2.2 Transmission .................................................................................................... 12
      1.2.3 Propagation ..................................................................................................... 14
      1.2.4 Reception ........................................................................................................ 15
      1.2.5 Measurements ............................................................................................... 15
      1.2.6 Sensitivities .................................................................................................... 18
  1.3 Common echosounder types ................................................................................... 21
      1.3.1 Single-beam ..................................................................................................... 21
      1.3.2 Split-beam ....................................................................................................... 22
      1.3.3 Sidescan .......................................................................................................... 23
      1.3.4 Multibeam ....................................................................................................... 23
      1.3.5 Wide-bandwidth ............................................................................................. 24
      1.3.6 Scanning .......................................................................................................... 24
      1.3.7 Acoustic Doppler ......................................................................................... 25
      1.3.8 Imaging ........................................................................................................... 25
  1.4 Common transducer platforms ................................................................................ 25
      1.4.1 Vessel ............................................................................................................. 25
      1.4.2 Autonomous moving .................................................................................... 27
      1.4.1 Autonomous stationary ................................................................................ 28
  1.5 Calibration methods ............................................................................................... 28
      1.5.1 Component ...................................................................................................... 29
      1.5.2 System ............................................................................................................. 31

2 Sphere calibration ........................................................................................................ 33
  2.1 Introduction ............................................................................................................. 33
      2.1.1 Equipment ....................................................................................................... 33
      2.1.2 Calibration procedure ................................................................................... 38
      2.1.3 Alternative platforms .................................................................................... 48

3 Calibration uncertainty ............................................................................................... 50
  3.1 Introduction ............................................................................................................. 50
  3.2 Random error ......................................................................................................... 50
  3.3 Systematic error ..................................................................................................... 51
      3.3.1 Sphere target strength ..................................................................................... 51
3.3.2 Sound speed and absorption .................................................. 54
3.3.3 Directivity pattern .................................................................. 55
3.3.4 Equivalent beam angle ............................................................ 56
3.3.5 Linearity and dynamic range .................................................... 56
3.3.6 System stability ....................................................................... 56
3.4 Total error ................................................................................... 57
3.4.1 Backscattering cross section ................................................... 57
3.4.2 Volume backscattering coefficient .......................................... 58

4 Calibration protocols .................................................................. 60
4.1 Simrad EK60, ES60, and ES70 .................................................. 60
  4.1.1 Introduction ............................................................................ 60
  4.1.2 Split-beam EK60 ................................................................. 66
  4.1.3 Split-beam ES60 and ES70 ..................................................... 71
  4.1.4 Single-beam ES60 and ES70 ............................................... 71
  4.1.5 Measurement error .............................................................. 72
  4.1.6 Documentation .................................................................... 73
  4.1.7 Quick start ........................................................................... 74
4.2 Simrad ME70 ............................................................................ 75
  4.2.1 Introduction ............................................................................ 75
  4.2.2 Theory .................................................................................. 75
  4.2.3 Calibration ............................................................................ 78

5 Emerging protocols .................................................................... 86
5.1 Multibeam echosounders ............................................................ 86
  5.1.1 Requirements ........................................................................ 86
  5.1.2 Measurements ...................................................................... 86
  5.1.3 System performance .............................................................. 87
  5.1.4 Seabed reference ................................................................... 87
  5.1.5 Sphere calibration ................................................................. 88
  5.1.6 Simrad SM20 ...................................................................... 90
5.2 Wide-bandwidth echosounders ................................................... 94
5.3 Partial-wave calibration .............................................................. 97
5.4 Acoustic Doppler current profilers ............................................. 100
  5.4.1 Component calibration .......................................................... 101
  5.4.2 Echosounder comparison ...................................................... 102
  5.4.3 Net catch comparison .......................................................... 103
5.5 Imaging sonars ......................................................................... 106
  5.5.1 DIDSON ............................................................................... 106
  5.5.2 Backscatter calibration ......................................................... 107
  5.5.3 Position calibration ............................................................... 107
5.6 Scanning sonar ......................................................................... 109
  5.6.1 Simrad SH80 ...................................................................... 110
Acoustic instrument calibration is fundamental to the quantitative use of its data for estimating aquatic resource abundance. Regular calibrations also allow instrument performance to be monitored to detect changes due to the environment or component dynamics, degradation, or failure.

This is the second ICES Cooperative Research Report (CRR) focussed on calibrations of acoustic instruments. The first, CRR No. 144 (Foote et al., 1987), was published during the era of analogue electronics more than a quarter of a century ago. Since then, not only has the acoustic equipment improved vastly with digital electronics and signal processing, but the techniques for applying them to studies of marine organisms have both advanced and diversified. Motivating, facilitating, and expediting these developments is the work of the Fisheries Acoustics, Science and Technology Working Group (WGFAST) of the International Council for the Exploration of the Sea (ICES).

CRR No. 144 guided the fisheries acoustics community to uniformly apply the sphere method to calibrate survey equipment, generally single-frequency, split-beam echosounders. Today, surveys of fishery resources are conducted using a large variety of acoustic instruments including, but not limited to, single-frequency, multifrequency, single-beam, split-beam, broad bandwidth, and multibeam echosounders; side-scan and scanning sonars; acoustic Doppler current profilers; and acoustic cameras. These instruments differ in the ways in which they function, are utilized, and the types of measurements they provide. In most cases, they also require different calibration techniques for optimizing the accuracy and characterizing the precision of the measurements.

With technological innovation proceeding at an ever faster pace, the challenge to create a comprehensive and practical guide to calibrating acoustic instruments is formidable. Obviously, not all acoustic instrumentation and methods are addressed here. The ones that are addressed are in various states of maturity. Therefore, the practical aims of this CRR are to document (i) acoustic instruments currently used in fisheries research and surveys, (ii) theoretical principles of calibrating these instruments, and (iii) methods currently being practiced for a selection of commonly used instruments.

To meet these goals, the WGFAST formed the Study Group on Calibration of Acoustic Instruments (SGCal) at its meeting in April 2009. The SGCal first met in San Diego, CA, USA in April 2010 to outline the document. Some chapters were drafted intersessionally. The SGCal met for the second time in Reykjavik, Iceland in May 2011 to collectively review some draft chapters. The drafts were refined intersessionally and merged. The draft CRR was collectively reviewed at meetings of the SGCal, in Pasaia, Spain in April 2013 and in New Bedford, MA, USA in May 2014. Multiple independent reviewers provided input, and the final editing was completed in 2014. The authors hope that this CRR will be a valuable reference to both novice and experienced users of fishery acoustic instruments, but recognize that it is a provisional guide that requires refinement and update as the field continues to progress.

David A. Demer
Chair, SGCal
Terms, symbols, and units

The names, symbols, and units of physical quantities must be precisely defined to assure effective scientific communication. However, different fields of study, e.g. physics, electrical engineering, radar, and sonar, often use different terms and symbols, and the same symbols may be used for different things, even within the same field. Furthermore, people generally have preferences for specific terms and symbols. Because universal consistency is not achievable, the aim for terms and symbols in this document is to define them uniquely and use them consistently.

The following terminology, based mostly on MacLennan et al. (2002), follows the Le Système International d’Unités (SI system). Symbols should uniquely represent a term. All symbols for variables should be italicized. Any symbol for a variable (x) that is not logarithmically transformed should be lower case. Any symbol for a logarithmically transformed variable, e.g. \( X = 10 \log_{10}(x/x_{\text{ref}}) \), with units of decibels referred to \( x_{\text{ref}} \) (dB re \( x_{\text{ref}} \)), should be capitalized. Deviations from these rules should be noted.

<table>
<thead>
<tr>
<th>Term</th>
<th>Symbol</th>
<th>Unit</th>
<th>Description</th>
</tr>
</thead>
<tbody>
<tr>
<td>Water</td>
<td>( t_w )</td>
<td>Degree Celsius (°C)</td>
<td>Heat, or average kinetic energy of particles in water.</td>
</tr>
<tr>
<td>salinity</td>
<td>( s_w )</td>
<td>Practical salinity unit (psu)</td>
<td>The total amount of dissolved material in water.</td>
</tr>
<tr>
<td>depth</td>
<td>( d_w )</td>
<td>Metre (m)</td>
<td>The vertical distance below the water surface.</td>
</tr>
<tr>
<td>pressure</td>
<td>( p_w )</td>
<td>Pascal (Pa) (= 10(^{-4}) dbar)</td>
<td>The force per unit area in water.</td>
</tr>
<tr>
<td>density</td>
<td>( \rho_w )</td>
<td>Kilogramme per cubic metre (kg m(^{-3}))</td>
<td>The mass density of water.</td>
</tr>
<tr>
<td>Voltage</td>
<td>( u_e )</td>
<td>Volt (V)</td>
<td>The square root of the mean of the squares (rms) of time-varying electric potential.</td>
</tr>
<tr>
<td>Current</td>
<td>( i_e )</td>
<td>Ampere (A)</td>
<td>The rms electric current.</td>
</tr>
<tr>
<td>impedance</td>
<td>( z_e )</td>
<td>Ohm (Ω)</td>
<td>The ability of a material to oppose the passage of an alternating electric current.</td>
</tr>
<tr>
<td>power</td>
<td>( p_e )</td>
<td>Watt (W) or VA or ( V^2 \Omega ) or ( A^2 \Omega )</td>
<td>The rms electrical energy per unit time.</td>
</tr>
<tr>
<td>Water sound</td>
<td>( c_w )</td>
<td>Metre per second (m s(^{-1}))</td>
<td>The distance sound travels per unit time in water.</td>
</tr>
<tr>
<td>speed</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>acoustic</td>
<td>( z_{aw} )</td>
<td>Kg m(^2) s(^{-1})</td>
<td>The product of ( \rho_w ) and ( c_w ).</td>
</tr>
<tr>
<td>impedance</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>frequency</td>
<td>( f )</td>
<td>Hertz (Hz) = cycles s(^{-1})</td>
<td>The number of complete cycles of a periodic wave per unit time.</td>
</tr>
<tr>
<td>Acoustic</td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>Term</td>
<td>Symbol</td>
<td>Unit</td>
<td>Description</td>
</tr>
<tr>
<td>-------------------------------------------</td>
<td>--------</td>
<td>------</td>
<td>-----------------------------------------------------------------------------</td>
</tr>
<tr>
<td>Acoustic frequency bandwidth</td>
<td>$b_f$</td>
<td>Hz</td>
<td>The difference between the highest and lowest $f$ in a signal or device.</td>
</tr>
<tr>
<td>Acoustic wavelength</td>
<td>$\lambda$</td>
<td>m</td>
<td>The distance spanned by one cycle of a periodic pressure wave.</td>
</tr>
<tr>
<td>Acoustic wavenumber</td>
<td>$k$</td>
<td>m$^{-1}$</td>
<td>The number of periodic wavelengths per $2\pi$-unit distance.</td>
</tr>
<tr>
<td>Pulse duration</td>
<td>$\tau$</td>
<td>s</td>
<td>The duration of a signal pulse.</td>
</tr>
<tr>
<td>Effective pulse duration</td>
<td>$\tau_{eff}$</td>
<td>s</td>
<td>The duration of a square shaped pulse that has the same energy as the echosounder-pulse shape after reception processing.</td>
</tr>
<tr>
<td>Range</td>
<td>$r$</td>
<td>m</td>
<td>The direct-path distance between objects, e.g., the transducer and the target.</td>
</tr>
<tr>
<td>Reference range</td>
<td>$r_0$</td>
<td>m</td>
<td>The $r$ from an acoustic source or target to which measurements are referred, conventionally $1$ m.</td>
</tr>
<tr>
<td>Acoustic pressure</td>
<td>$p$</td>
<td>Pa</td>
<td>The rms deviation of local pressure from the ambient.</td>
</tr>
<tr>
<td>Reference acoustic pressure</td>
<td>$p_{ref}$</td>
<td>Pa</td>
<td>The reference $p$, conventionally $1\mu$Pa for underwater sound.</td>
</tr>
<tr>
<td>Acoustic power</td>
<td>$p_a$</td>
<td>W</td>
<td>The rms acoustic energy per unit time.</td>
</tr>
<tr>
<td>Acoustic intensity</td>
<td>$i$</td>
<td>W m$^{-2}$</td>
<td>The $i$ at $r$.</td>
</tr>
<tr>
<td>Acoustic intensity at $r_0$</td>
<td>$i_0$</td>
<td>W m$^{-2}$</td>
<td>The $i$ of a plane wave with rms $p = p_{ref}$.</td>
</tr>
<tr>
<td>Spherical spreading loss</td>
<td>No symbol</td>
<td>m$^{-2}$</td>
<td>The reduction in $i$ with $r$ resulting from spherical expansion of the wavefront.</td>
</tr>
<tr>
<td>Absorption loss</td>
<td>No symbol</td>
<td>m$^{-1}$</td>
<td>The reduction in $i$ with $r$ resulting from conversion to heat.</td>
</tr>
<tr>
<td>Absorption coefficient</td>
<td>$\alpha_a$</td>
<td>dB m$^{-1}$</td>
<td>A metric of absorption loss.</td>
</tr>
<tr>
<td><strong>Transduction</strong></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>Transmit voltage</td>
<td>$u_{et}$</td>
<td>V</td>
<td>The rms $u_e$ input to a transducer.</td>
</tr>
<tr>
<td>Transmit current</td>
<td>$i_{et}$</td>
<td>A</td>
<td>The rms $i_e$ input to a transducer.</td>
</tr>
<tr>
<td>Transducer electrical impedance</td>
<td>$z_{et}$</td>
<td>$\Omega$</td>
<td>The ability of a transducer to oppose the passage of an alternating electric current.</td>
</tr>
<tr>
<td>Receiver electrical impedance</td>
<td>$z_{er}$</td>
<td>$\Omega$</td>
<td>The ability of an echosounder receiver to oppose the passage of an alternating electric current.</td>
</tr>
<tr>
<td>Term</td>
<td>Symbol</td>
<td>Unit</td>
<td>Definition</td>
</tr>
<tr>
<td>--------------------------------</td>
<td>--------</td>
<td>----------</td>
<td>---------------------------------------------------------------------------</td>
</tr>
<tr>
<td>Transmit electric power</td>
<td>$p_{et}$</td>
<td>W</td>
<td>The electrical power input to a transducer.</td>
</tr>
<tr>
<td>Transmit acoustic power</td>
<td>$p_{ac}$</td>
<td>W</td>
<td>The acoustic energy per unit time output from a transducer.</td>
</tr>
<tr>
<td>Transducer efficiency</td>
<td>$\eta$</td>
<td>Dimensionless</td>
<td>The proportion of $p_{et}$ converted to $p_{ac}$.</td>
</tr>
<tr>
<td>Transmit pressure</td>
<td>$p_t$</td>
<td>Pa</td>
<td>The $p$ output from a transducer.</td>
</tr>
<tr>
<td>Directional angles</td>
<td>$\alpha, \beta$</td>
<td>Degree (°)</td>
<td>The angle coordinates in orthogonal planes, typically alongships and athwartships or aligned with the major and minor transducer axes, respectively.</td>
</tr>
<tr>
<td>Gain</td>
<td>$g(\alpha, \beta)$</td>
<td>Dimensionless or Decibel referred to 1 (dB re 1)</td>
<td>The ratio of $i$ values, observed at a distant point, resulting from transmissions, with constant $p_{et}$ of a real transducer, and an idealized lossless omnidirectional transducer. Gain accounts for losses referred to a point on the electrical side of the transducer.</td>
</tr>
<tr>
<td>On-axis gain</td>
<td>$g_0$</td>
<td>Dimensionless or dB re 1</td>
<td>The $g(\alpha, \beta)$ on the transducer beam axis ($\alpha = \beta = 0$).</td>
</tr>
<tr>
<td>Transducer directivity</td>
<td>$d(\alpha, \beta)$</td>
<td>Dimensionless or dB re 1</td>
<td>The one-way directional gain of a real transducer, omitting losses.</td>
</tr>
<tr>
<td>Transducer directivity pattern</td>
<td>$b(\alpha, \beta)$</td>
<td>Dimensionless or dB re 1</td>
<td>The ratio of $d(\alpha, \beta)$ and its maximum value.</td>
</tr>
<tr>
<td>Transducer beamwidth</td>
<td>$\theta_{-3 \text{ dB}}$</td>
<td>°</td>
<td>The angle from the transducer beam axis to the point of half-power, in the $\theta$, $\alpha$, or $\beta$ planes.</td>
</tr>
<tr>
<td>Transducer beamwidth offset</td>
<td>$\alpha_0$ or $\beta_0$</td>
<td>°</td>
<td>The angular displacement of the transducer maximum-response axis beam from the axis of zero split-beam phase, in the $\alpha$ or $\beta$ planes.</td>
</tr>
<tr>
<td>Transducer angle sensitivity</td>
<td>$A_\alpha$ or $A_\beta$</td>
<td>Electrical°/geometric°</td>
<td>A factor to convert split-beam electrical angles to target bearing angles in the $\alpha$ or $\beta$ planes.</td>
</tr>
<tr>
<td>Source intensity or Source level</td>
<td>$i_s$ or $SL$</td>
<td>W m$^{-2}$ or dB re 1 $\mu$Pa at $r_0$</td>
<td>The $i_s$ in the direction $(\alpha, \beta)$ of the target.</td>
</tr>
<tr>
<td>Transmit intensity</td>
<td>$i_t$</td>
<td>W m$^{-2}$</td>
<td>The intensity of a plane wave transmitted from a real transducer.</td>
</tr>
<tr>
<td>Parameter</td>
<td>Symbol</td>
<td>Unit</td>
<td>Definition</td>
</tr>
<tr>
<td>--------------------------------------------------------------------------</td>
<td>--------</td>
<td>---------------</td>
<td>---------------------------------------------------------------------------</td>
</tr>
<tr>
<td>Omni-directional transmit acoustic intensity</td>
<td>$i_{\text{omni}}$</td>
<td>W m$^{-2}$</td>
<td>The $i$ for an idealized lossless omni-directional transducer.</td>
</tr>
<tr>
<td>Equivalent two-way beam angle</td>
<td>$\psi$</td>
<td>steradian (sr)</td>
<td>The solid angle subtended by an ideal conical beam that would produce the same volume integral as the square of the normalized transducer directivity.</td>
</tr>
<tr>
<td>Effective receiving area</td>
<td>$a_{r}(\alpha, \beta)$</td>
<td>m$^2$</td>
<td>The area of a real transducer available to receive acoustic power and transfer it to a matched load (i.e. $z_p = z_s$), referred to the same point on the electrical side of the transducer as $g_0$.</td>
</tr>
<tr>
<td>Received electric power</td>
<td>$p_{\text{er}}$</td>
<td>W or dB re 1 W</td>
<td>The $p_e$ output from a transducer, referred to the same point as $g_s$ and $v_r$.</td>
</tr>
<tr>
<td>On-axis received power</td>
<td>$p_{\text{er0}}$ or $p_{\text{er}10}$</td>
<td>W or dB re 1 W</td>
<td>The $p_e$ in the direction $(\alpha, \beta)$ of the maximum $b(\alpha, \beta)$.</td>
</tr>
<tr>
<td>Transmit voltage sensitivity</td>
<td>$s_{u}$ or $S_u$</td>
<td>$\mu$Pa V$^{-1}$ or dB re 1 $\mu$Pa V$^{-1}$</td>
<td>The $SL$ resulting from a 1-V rms sinusoidal signal applied to the transducer terminals.</td>
</tr>
<tr>
<td>Transmit current sensitivity</td>
<td>$s_{i}$ or $S_i$</td>
<td>$\mu$Pa A$^{-1}$ or dB re 1 $\mu$Pa A$^{-1}$</td>
<td>The $SL$ resulting from a 1-A rms sinusoidal signal applied to the transducer terminals.</td>
</tr>
<tr>
<td>Receive voltage sensitivity</td>
<td>$m_{u}$ or $M_u$</td>
<td>V $\mu$Pa$^{-1}$ or dB re 1 V $\mu$Pa$^{-1}$</td>
<td>The unloaded $\mu$ at a point on the electrical side of the transducer resulting from $i_{ref}$ applied to the transducer surface.</td>
</tr>
<tr>
<td>Reciprocity parameter</td>
<td>$j$</td>
<td>dimensionless</td>
<td>The ratio of the response of a linear, passive, reversible electroacoustic transducer acting as a receiver, $m_{w}$, to its response as a transmitter, $s_{i}$.</td>
</tr>
<tr>
<td>Sphere</td>
<td>$\rho_s$</td>
<td>kg m$^{-3}$</td>
<td>Mass density of an elastic sphere.</td>
</tr>
<tr>
<td>Sphere compressional wave sound speed</td>
<td>$c_c$</td>
<td>m s$^{-1}$</td>
<td>The compressional (or longitudinal) wave sound speed of an elastic sphere.</td>
</tr>
<tr>
<td>Sphere shear wave sound speed</td>
<td>$c_s$</td>
<td>m s$^{-1}$</td>
<td>The shear (or transverse) wave sound speed of an elastic sphere.</td>
</tr>
<tr>
<td>Metrics</td>
<td>$r_{mn}$</td>
<td>Dimensionless</td>
<td>The quotient of signal and noise power.</td>
</tr>
<tr>
<td>Spherical scattering cross-section</td>
<td>$\sigma_{sp}$</td>
<td>m$^2$</td>
<td>The area of an acoustic target effectively scattering acoustic power.</td>
</tr>
<tr>
<td>Backscattering cross-section</td>
<td>$\sigma_{bs}$</td>
<td>m$^2$</td>
<td>The area of an acoustic target effectively backscattering acoustic power, $\sigma_{bs} = \sigma_{sp}/4\pi$, at $r_0$.</td>
</tr>
</tbody>
</table>
Calibration of acoustic instruments

Target strength

<table>
<thead>
<tr>
<th>Sampled area</th>
<th>$a$</th>
<th>m$^2$</th>
<th>The area contributing to a received signal.</th>
</tr>
</thead>
<tbody>
<tr>
<td>Surface backscattering coefficient or Surface backscattering strength</td>
<td>$s_x$</td>
<td>m$^2$ m$^{-2}$</td>
<td>The backscattering cross-section per unit of surface area, assuming the area is the intersection of $\psi$ and a sphere with radius $r$ centred on the transducer.</td>
</tr>
<tr>
<td>Sampled volume</td>
<td>$v$</td>
<td>m$^3$</td>
<td>The volume contributing to a received signal.</td>
</tr>
<tr>
<td>Volume backscattering coefficient or Volume backscattering strength</td>
<td>$s_v$</td>
<td>m$^2$ m$^{-3}$</td>
<td>The sum of $s_v$ per unit of water volume.</td>
</tr>
<tr>
<td>Survey area</td>
<td>$a_s$</td>
<td>m$^2$</td>
<td>The area of the survey region.</td>
</tr>
<tr>
<td>Area backscattering coefficient or Nautical area backscattering coefficient</td>
<td>$s_a$</td>
<td>m$^2$ m$^{-2}$</td>
<td>The integral of $s_a$ over a range of depths; or $s_a$ multiplied by a scaling factor $(4\pi(1852)^2$ nautical mile$^2$ m$^{-2}$).</td>
</tr>
<tr>
<td>Scatterer volume density</td>
<td>$\rho_v$</td>
<td>Number m$^{-3}$</td>
<td>The number of scatterers per unit of sampled volume.</td>
</tr>
<tr>
<td>Scatterer areal density</td>
<td>$\rho_a$</td>
<td>Number m$^{-2}$</td>
<td>The number of scatterers per unit of sampled area.</td>
</tr>
<tr>
<td>Scatterer abundance</td>
<td>$n_b$</td>
<td>Number</td>
<td>The number of scatterers within a sampled range and area.</td>
</tr>
<tr>
<td>Scatterer biomass</td>
<td>$m_b$</td>
<td>kg</td>
<td>The mass of scatterers within a sampled range and area.</td>
</tr>
</tbody>
</table>
1 Introduction

1.1 Scope and outline

Acoustic sampling has long been a standard survey tool for estimating the abundance and distribution of fish, zooplankton, and their seabed habitat (Kimura, 1929; Sund, 1935; Holliday, 1972a; Nielson et al., 1980). Typically, acoustic surveys are conducted using multifrequency echosounders that transmit sound pulses down beneath the ship and receive echoes from animals and the seabed in the path of the sound waves (Simmonds and MacLennan, 2005). Generally, for surveys of animals, the backscatter signal is normalized to the range-dependent observational volume yielding the volume backscattering coefficient, which provides indications of the target type and behaviour. Objects scatter sound if their product of mass density and sound speed is different from that of the surrounding medium. A fish with a swimbladder has a large acoustic-impedance contrast (Foote, 1980), and thus has a large reflecting area, backscattering cross-section. Plankton, e.g. krill and salps, generally have much lower acoustic-impedance contrasts, but can produce large volume backscattering coefficients when they are aggregated in large densities (Hewitt and Demer, 1991, 2000). Under certain conditions, the summed and averaged volume backscattering coefficients are linearly related to the density of the fish or plankton aggregations that contributed to the echoes (Foote, 1983a). The number density can be estimated by dividing the integrated volume backscattering coefficient from an aggregation of target species by the average backscattering cross section from a representative animal (Ehrenberg and Lytle, 1972). An estimate of animal abundance is then obtained by multiplying the average estimated fish density and the survey area.

Increasingly, multifrequency echosounder surveys are being augmented with samples from other acoustic instruments such as multibeam echosounders (Gerlotto et al., 1999; Simmonds et al., 1999; Berger et al., 2009; Colbo et al., 2014), multibeam imaging sonars (Korneliussen et al., 2009; Patel and Ona, 2009), and long-range scanning sonars (Bernasconi et al., 2009; Nishimori et al., 2009; Stockwell et al., 2013)(Figure 1.1). Use of these instruments provides information on many more aspects of the biotic and abiotic environment, e.g. bathymetry, seabed classification (Humorstad et al., 2004; Cutter and Demer, 2014), oceanographic fronts (Wade and Heywood, 2001), mixed-layer depths, anoxic regions, internal waves (Lavery et al., 2010a), turbulence (Stanton et al., 1994), currents, and methane seeps, all contributing to a broader ecosystem perspective (Demer et al., 2009a). In each case, the quantitative use of the data requires that the acoustic instrument is calibrated.
Instrument calibration involves the characterization of measurement accuracy (bias or systematic error) and precision (variability or random error). Sampling with the calibrated instrument involves additional systematic and random error (Demer, 2004). Calibration accuracy is estimated and optimized by comparing measured and assumed values for a standard, and correcting for the difference. The selection and characterization of a calibration standard is, therefore, paramount to the accuracy of an instrument calibration (Foote and MacLennan, 1984). Calibration precision is estimated by comparing multiple measures of a standard. Importantly, the performance of an instrument and thus its calibration accuracy and precision may change vs. time or the environment (Demer and Hewitt, 1993; Brierley et al., 1998a; Nam et al., 2007). Therefore, instruments should be calibrated frequently within the range of environments where they are used to make calibrated measurements (Demer and Renfree, 2008). If this is not possible, account should be made for any changes in the instrument or environment that appreciably affect the calibration accuracy and precision.

This report includes general instruction and current best practices for calibrating a selection of acoustic instruments commonly used to conduct fishery science and surveys. It also describes some less developed protocols for other acoustic instruments. For practical reasons, not all fishery acoustic instruments are included.

The remainder of Chapter 1 (i) summarizes some of the theoretical principles of acoustic instruments used to conduct fishery research and surveys, (ii) describes some commonly used instruments and their deployment platforms, and (iii) briefly introduces some common methods for calibrating acoustic instruments. Readers seeking only protocols for calibrating echosounders may wish to skip this and other sections and consult the Contents table to access information related to their interest and need.

Chapter 2, details the sphere calibration method. Chapter 3 explores uncertainty in sphere calibrations. Chapter 4 describes protocols for calibrating some commonly used echosounders. Chapter 5 describes emerging protocols for some other acoustic instruments. Chapter 6 acknowledges valuable contributions to this CRR by people not included in the list of authors.

1.2 Echosounder theory

The theory of sound transduction, propagation, and scattering is documented in numerous books and journal articles (see References). It is based on the same principles as radar theory (Rihaczek, 1969; Kerr, 1988), which is used for calibration and data processing in many commonly used acoustic instruments. Section 1.2.1 documents radar theory as it relates to these instruments; Section 1.2.2 highlights the calibration param-

Figure 1.1. A conceptual image of acoustic-sampling beams projecting from a survey vessel equipped with multifrequency split-beam (green) and multibeam (orange) echosounders, multibeam imaging sonar (purple), and long-range scanning sonar (grey).
12

ICES Cooperative Research Report No. 326

eters; Section 1.2.3 summarizes the measurements that are typically derived from echosounder data; and Section 1.2.4 reconciles this radar theory with some aspects of sonar theory used by other systems. The content of these sections is largely a synthesis of Urick (1983), Kerr (1988), Simrad (1993), Medwin and Clay (1998), Bodholt (2002), Andersen (2006), and other references cited in the text. For simplicity and clarity, the theory is mostly presented in the linear domain, i.e., not in decibels.

1.2.1 Introduction

An echosounder generates, amplifies, and transmits an acoustic pulse (Figure 1.2), which propagates outward. When the pulse reaches a target object, e.g., zooplankton, fish, or the seafloor, the interaction results in a reflected or scattered wave. Characteristics of the target modulate the scattered wave amplitude, shape, and frequency, and these changes may provide remote-sensing information. Part of this wave travels back to the echosounder where it is converted to voltage, amplified, filtered to remove noise, digitized, and thus received as backscatter (Furusawa, 1991). Real sonar systems may also include other steps such as heterodyning and demodulating. The delay between the pulse transmission and the backscatter reception is converted into an estimate of the target range using an estimate of the harmonic mean sound speed along the propagation path.

![Figure 1.2. Diagram of a notional echosounder. Although a separate transmitter (projector) and receiver (“hydrophone”) are shown (i.e., a bistatic system), most echosounders use a single transducer to transmit and receive (i.e., a monostatic system). The analogue-to-digital converter (ADC) digitizes the received, amplified and filtered voltage signal.](image)

1.2.2 Transmission

1.2.2.1 Transmit power

A narrow bandwidth echosounder outputs sinusoidal pulses of mean transmit electric power, $p_{et}$ (W), defined by the root-mean-square (rms) transmit voltage, $u_{et}$ (V), and rms transmit current, $i_{et}$ (A), applied to an electro-acoustic transducer with transducer electrical impedance, $z_{et}$ (Ω). These variables are complex (having amplitude and phase), and are related by Ohm’s Law:

$$p_{et} = \frac{u_{et}^2}{z_{et}} = i_{et}^2 z_{et}.$$  

(1.1)

The transmit energy, equal to $p_{et}$ integrated over time, increases with the pulse duration, $\tau$ (s). The resulting acoustic signal has a frequency bandwidth, $b_f \equiv 1/\tau$ (Hz), about the signal frequency, $f$ (Hz). In water with sound speed, $c_w$ (m s$^{-1}$), the acoustic wavelength, $\lambda$ (m), of the signal is:
\[ \lambda = \frac{c_w}{f} \]  

(1.2)

Values for \( c_w \) and thus \( \lambda \) vary during the calibration and subsequent use of the echo-sounder, and modulate calibration accuracy and precision.

### 1.2.2.2 Intensity

Analogous to Ohm’s law (Equation 1.1), the mean transmit acoustic intensity, \( i_t \) (W m\(^{-2}\)), of the plane wave is equal to the square of the rms transmit acoustic pressure, \( p_t \) (Pa = N m\(^{-2}\)), divided by the water acoustic impedance, \( z_{aw}(\text{kg m}^{-2} \text{s}^{-1}) \), which is the product of \( c_w \) and mass density, \( \rho_w \) (kg m\(^{-3}\)):

\[ i_t = \frac{p_t^2}{z_{aw}} = \frac{p_t^2}{\rho_w c_w} = \frac{p_{at}}{a}. \]  

(1.3)

By definition, \( i_t \) equals the mean transmit acoustic power, \( p_{at} \) (W), per unit area, \( a \) (m\(^2\)). Post-calibration variation in \( \rho_w \) values may also affect the calibration accuracy and precision.

The acoustic pressure of the resulting plane wave is often related to a reference acoustic pressure, \( (p_{ref} = 1 \, \mu\text{Pa}, \text{conventionally}) \). Thus, the reference acoustic intensity, \( i_{ref}(\text{W m}^{-2}) \), is:

\[ i_{ref} = \frac{p_{ref}^2}{\rho_w c_w}. \]  

(1.4)

Unlike \( p_{ref} \), values for \( i_{ref} \) may be modulated by variation in \( z_{aw} \).

### 1.2.2.3 Directivity

The one-way transducer directivity, \( d(\alpha, \beta) \) (dimensionless), describes the ratio of acoustic intensity, \( i(\alpha, \beta) \) (W m\(^{-2}\)), and \( i(\alpha, \beta) \) averaged over all directions, observed at a distant point, where \( \alpha \) (°) and \( \beta \) (°) are orthogonal “alongships” and “athwartships” angles. The function \( d(\alpha, \beta) \) is dependent on \( \lambda \), which changes with \( c_w \), following Equation (1.2). The one-way transducer directivity pattern, \( b(\alpha, \beta) \) (dimensionless) is the ratio of \( d(\alpha, \beta) \) and its maximum value (Figure 1.3).
Figure 1.3. Equivalent representations of transducer directivity pattern, \( b(\theta) \) (dimensionless) (left), and \( B(\theta) = 10 \log_{10} b(\theta) \) (dB re 1) (right), versus angle \( \theta \) off the beam axis \( (\theta = \alpha, \beta) \) of a piston transducer with a diameter equal to 50 wavelengths.

1.2.2.4 Gain

The transducer gain, \( g \) (dimensionless), is defined as the ratio of \( i(\alpha, \beta) \) from a real transducer and an idealized lossless omnidirectional transducer, \( i_{\text{omni}} \) (W m\(^{-2}\)). It accounts for the transducer efficiency, \( \eta \) (dimensionless), which is the proportion of \( p_{\text{et}} \), referred to a point on the electrical side of the transducer, converted to \( p_{\text{et}} \):

\[
g(\alpha, \beta) = \frac{i(\alpha, \beta)}{i_{\text{omni}}} = \frac{p_{\text{et}}(\alpha, \beta)}{p_{\text{et}}} = \eta d(\alpha, \beta) = g_0 b(\alpha, \beta) .
\]

(1.5)

The on-axis gain, \( g_0 \) (dimensionless), is the \( g(\alpha, \beta) \) in the direction of maximum \( d \) (usually at \( \alpha = 0^\circ, \beta = 0^\circ \)). Both \( g_0 \) and \( b(\alpha, \beta) \) are dependent on \( \lambda \), which changes with \( c_w \) (Bodholt, 2002).

1.2.3 Propagation

1.2.3.1 Attenuation

As the transmitted pulse propagates to range, \( r \) (m), \( p_{\text{et}} d(\alpha, \beta) \) is attenuated due to spreading of the wavefront (spherical spreading loss = \( 1/4\pi r^2 \)) and conversion to heat in the water (absorption loss = \( 10^{-\alpha_a r/10} \)):

\[
i = p_{\text{et}} d(\alpha, \beta) \frac{10^{-\alpha_a r/10}}{4\pi r^2} = p_{\text{et}} g(\alpha, \beta) \frac{10^{-\alpha_a r/10}}{4\pi r^2} = \eta d(\alpha, \beta) \frac{10^{-\alpha_a r/10}}{4\pi r^2} = g_0 b(\alpha, \beta) \frac{10^{-\alpha_a r/10}}{4\pi r^2} .
\]

(1.6)

where \( \alpha_a \) (dB m\(^{-1}\)) is the absorption coefficient. Values for \( \alpha_a \), and their uncertainty, increase with \( f \) and vary with water temperature, salinity, pressure, pH, and \( \lambda \) (Urick, 1983).

1.2.3.2 Source intensity

Referring \( i \) to a reference range, \( r_0 \) (m), conventionally 1 m, from the acoustic centre of the transducer in the direction of maximum \( d \) yields the source intensity, \( i_s \) (W m\(^{-2}\)):

\[
i_s = p_{\text{et}} g_0 \frac{10^{-\alpha_a r_0/10}}{4\pi r_0^2} = \eta d g_0 \frac{10^{-\alpha_a r_0/10}}{4\pi r_0^2} = i_{\text{et}} \eta d g_0 \frac{10^{-\alpha_a r_0/10}}{4\pi r_0^2} .
\]

(1.7)

The \( i_s \) is commonly expressed in decibels referred to the intensity of a plane wave with \( p = 1 \) \( \mu \)Pa \((p_{\text{ref}})\), and termed source level, \( \text{SL} \) (dB re \( p_{\text{ref}}^2 \) W m\(^{-2}\) at \( r_0 \)):

\[
\text{SL} = 10 \log_{10} \left( \frac{i_s}{p_{\text{ref}}^2} \right) = 10 \log_{10} \left( \frac{10^{-\alpha_a r_0/10}}{4\pi r_0^2} \right) + 10 \log_{10} \left( \frac{p_{\text{ref}}^2}{p_{\text{ref}}^2} \right) + 10 \log_{10} \left( \frac{p_{\text{ref}}^2}{p_{\text{ref}}^2} \right) .
\]

(1.8)

Assuming \( \rho_w = 1035 \) kg m\(^{-3}\), \( c_w = 1500 \) m s\(^{-1}\), and \( p_{\text{ref}} = 10^{-4} \) Pa,

\[
\text{SL} = 10 \log_{10} (i_s) + 181.91 = 10 \log_{10} \left( \frac{10^{-\alpha_a r_0/10}}{4\pi r_0^2} \right) + 181.91 .
\]

(1.9)
1.2.3.3 Backscattering cross section

When the transmitted wavefront encounters a target, the area effectively scattering the incident acoustic power is its spherical scattering cross section, \( \sigma_{sp} (m^2) \), which equals the backscattering cross section, \( \sigma_{bs} (m^2) \), multiplied by 4\( \pi \) (MacLennan et al., 2002). Backscattering cross section is also known as the differential backscattering cross section (Medwin and Clay, 1998). As the reflected pulse travels between the target and the transducer, \( i \) is again attenuated due to spreading and absorption losses:

\[
i = p_{et} g(\alpha, \beta) 10^{-a_{sr}/10} - 4\pi\sigma_{bs} 10^{-a_{sr}/10} = p_{et} g(\alpha, \beta) \sigma_{bs} 4\pi\sigma_{bs} 10^{-a_{sr}/5}.
\]

(1.10)

1.2.4 Reception

1.2.4.1 Effective receiving area

The effective receiving area, \( a_r (m^2) \), is the area of the transducer that transfers its to received electric power, \( p_{er} (W) \), referred to the same point on the electrical side of the transducer as \( p_{et} \) and \( g(\alpha, \beta) \) (Kerr, 1988):

\[
a_r = \frac{\lambda^2 g(\alpha, \beta)}{4\pi}.
\]

Therefore,

\[
p_{er} = ia_r = \frac{p_{et} g(\alpha, \beta) \sigma_{bs} \lambda^2 g(\alpha, \beta)}{4\pi 10^{-a_{sr}/5} 4\pi} = \frac{p_{et} g(\alpha, \beta) \lambda^2 g(\alpha, \beta) \sigma_{bs}}{16\pi^2 r^4 10^{-a_{sr}/5}}.
\]

(1.12a)

Values for \( p_{er} \) change with \( a_r \), which changes with \( \lambda \) and thus \( c_w \) (Bodholt, 2002).

1.2.4.2 Effective pulse duration

To increase the signal-to-noise ratio, \( r_{sn} \) (dimensionless), the transfer function of the receiver, including the transducer, receiver electronics, and digital filters, suppresses noise outside \( b_f \) and, in the process, delays, and distorts the echo pulse. Account must be made for this “filter delay” when estimating \( r \) from the sound propagation delay time, \( t \) (s), and \( c_w \) (see Simmonds and MacLennan, 2005). Correction for filter delay improves the accuracy of attenuation compensation, i.e. time-varied gain (Fernandes and Simmonds, 1996), and is especially critical for calibrations and measurements made at short ranges (Ona et al., 1996). Modern echosounders generally correct for filter delay.

The receiver filter also modifies the amplitude and duration of the pulse, potentially altering the signal energy, \( p_{er} \). The perceived signal energy, discussed in MacLennan (1981), equals the measured instantaneous received power, \( p_{er}(t) \) (W), a function of time, \( t \) (s), integrated over the period when it exceeds a threshold. For a single target, this period is the effective pulse duration, \( \tau_{eff} \) (s) (see Sections 1.2.5.1 and 1.2.5.3).

1.2.5 Measurements

1.2.5.1 Range

The range to a scatterer is estimated by multiplying half of the propagation delay by an estimate of \( c_w \) for the propagation path. The maximum resolution of the range measurement is approximately equal to \( c_w/2 b_f \) (Burdic, 1991), assuming that the bandwidth of the receiver is larger than the transmitted signal bandwidth, \( b_f \) (Hz), and the Doppler shift in the scattered return from the target is negligible. For gated continuous
wave (CW) signals with pulse duration, $\tau$ (s), $b_f \equiv 1/\tau$. For frequency-modulated signals, $b_f$ is larger and the range resolution is proportionately smaller.

### 1.2.5.2 Backscattering cross section

Values of $\sigma_{bs}$ may be calculated by rearranging Equation (1.12):

$$\sigma_{bs} = \frac{\text{per}16\pi^2 \cdot 10^9 a^{7/5}}{\text{pet}^2(a \beta)^2 \cdot a^2}.$$  

(1.12b)

Some echosounders approximate $p_{er}$ by the maximum measured received power, $\max[p_{er}(t)] \ (W)$, and account for any receiver gain or attenuation via the calibrated gain parameter (Simrad, 1993). Alternatively, $p_{er}$ may be approximated by $\int_{t_{eff}} p_{er}(t) \ dt/t_{eff}$. Measurements of $\sigma_{bs}$ are commonly presented in decibels referred to 1 m$^2$, termed target strength, TS (e.g. dB re 1 m$^2$).

For a target on the beam axis, $g_0 = g(\alpha = 0^\circ, \beta = 0^\circ)$ (dB re 1), is obtained by rearranging Equation (1.12b):

$$g_0 = \left(\frac{p_{er} \cdot 16\pi^2 \cdot 10^9 a^{7/5}}{\text{pet}^2(a \beta)^2 \cdot a^2} \right)^{0.5} .$$

(1.12c)

The on-axis gain is the principle parameter estimated from a calibration, typically using a standard target (see Chapter 2).

In a self-reciprocity calibration (see Section 1.5.2.1), the transmitted acoustic pulse is received from a near-perfect reflector (e.g. a still air–water interface). Attenuation is calculated for one-way propagation from the transducer image. The image range, $r \ (m)$, is twice the actual range from the transducer to the reflector, $r (n = 2r)$:

$$p_{er} = p_{et} g_0 \frac{10^{-a_r/10} \lambda^2 \theta_0}{4\pi r^2} = \frac{p_{et} \lambda^2 \theta_0^2}{64\pi^2 r^2 10^9 a^{7/5}} .$$

(1.13a)

Rearranging:

$$g_0 = \left(\frac{p_{et} \cdot 64\pi^2 \cdot 10^9 a^{7/5}}{\text{pet}^2 \cdot a^2} \right)^{0.5} ,$$

(1.13b)

which is independent of $\sigma_{bs}$. As self-reciprocity calibrations are typically made at short ranges, it is critical for estimates of $r$ to be corrected for filter delay (see Simmonds and MacLennan, 2005) or measured physically.

### 1.2.5.3 Surface backscattering

The equivalent two-way beam angle, $\psi \ (sr)$, is the solid angle subtended by an ideal conical beam that would produce the same volume integral as the square of the normalized real transducer directivity pattern. The area contributing to a received signal is the sampled area, $a_s \ (m^2)$. The surface backscattering coefficient, $s_s \ (m^2 \cdot m^{-2})$, is the $\sigma_{bs}$ per unit of $a_s$. Assuming $a_s$ corresponds to the intersection of $\psi$ and a sphere with radius equal to range $r$, centred on the transducer:

$$S_s = \frac{\delta \sigma_{bs}}{\delta a_s} = \frac{p_{er} 16\pi^2 \cdot 10^9 a^{7/5}}{\text{pet}^2(a \beta)^2 \cdot 2 \cdot \psi} .$$

(1.14)
This definition of $s_s$ approximates normal incidence scattering from a flat seabed if the acoustic beam is narrow. Values of $s_s$ are commonly expressed in decibels referred to $1 \text{ m}^2 \text{ m}^{-2}$, and termed *Surface backscattering strength*, $S_s$ (dB re $1 \text{ m}^2 \text{ m}^{-2}$).

### 1.2.5.4 Volume backscattering

The *sample volume*, $v_s$ (m$^3$), is the volume of water contributing to a received signal, corresponding to the intersection of $\psi$ and a sphere of thickness $c_w \tau / 2$ (Foote, 1991a). *Volume backscattering coefficient*, $s_v$ (m$^2$ m$^{-3}$), is the total backscattering cross section per unit of sample volume:

$$ s_v = \frac{\delta s_b}{\delta v_s} = \rho_b \sigma_{bs} = \frac{p_{er} 32 \pi r^2 10^{-6} a^2 \psi}{\rho_{et} \beta \pi^2 c_w \tau \psi a_{corr}^2} \text{ (1.15)} $$

where $\rho_b$ (number m$^{-3}$) is the scatterer *volume density*, and $\sigma_{bs}$ is the mean backscattering cross section for the scatterers in the sampled volume. To account for the effects of the receiver filter, replace $\tau$ with $\tau_{eff}$:

$$ s_v = \frac{p_{er} 32 \pi r^2 10^{-6} a^2 \psi}{\rho_{et} \beta \pi^2 c_w \tau \psi a_{corr}^2} \text{ (1.16)} $$

where $s_{a \ corr} = \left( \tau_{eff} / \tau \right)^{0.5}$ (dimensionless) is the *scatterer correction factor*. $s_{a \ corr}^2$ is squared in Equation (1.16) so that it relates linearly to gain. During calibration with a target near the beam axis, $s_{a \ corr}^2$ can be estimated from the ratio of the energies in the perceived signal, and a square pulse defined by $\max[p_{er}(t)]$ and $\tau$:

$$ s_{a \ corr}^2 = \frac{\tau_{eff}}{\tau} = \frac{\int_{\tau_{eff}}^{\tau} p_{er}(t) dt}{\max[p_{er}(t)] \tau} \text{ (1.17)} $$

Values of $s_v$ are commonly expressed in decibels referred to $1 \text{ m}^2 \text{ m}^{-3}$, and termed *volume backscattering strength*, $S_v$ (dB re $1 \text{ m}^2 \text{ m}^{-3}$).

### 1.2.5.5 Area backscattering

*Area backscattering coefficient*, $s_a$ (m$^2$ m$^{-2}$), is the integral of $s_v$ over a measurement range, $r_1$ to $r_2$, often associated with a sampled depth range.

$$ s_a = \int_{r_1}^{r_2} s_v dr \text{ (1.18a)} $$

*Nautical area backscattering coefficient*, $s_A$ (m$^2$ nautical mile$^{-2}$), is the same variable with different units and a legacy factor ($4 \pi$) (MacLennan et al., 2002).

$$ s_A = 4 \pi (1852)^2 \int_{r_1}^{r_2} s_v dr \text{ (1.18b)} $$

### 1.2.5.6 Areal density

*Scatterer areal density*, $\rho_a$ (number m$^{-2}$) or $\rho_A$ [number (nautical mile)$^{-2}$], is the number of scatterers within the measurement range, e.g. depth range, per unit of sampled area, e.g. sea surface area:

$$ \rho_a = \frac{s_a}{\sigma_{bs}}, \text{ and } \rho_A = \frac{s_A}{4 \pi \sigma_{bs}} \text{ (1.19a)} $$

$$ \rho_a = \frac{s_a}{\sigma_{bs}}, \text{ and } \rho_A = \frac{s_A}{4 \pi \sigma_{bs}} \text{ (1.19b)} $$
1.2.5.7 Abundance

Scatterer abundance, \( n_b \) (number), is the number of target organisms within the measurement range, e.g. depth range, and the sampled area, \( a_s \) (m\(^2\)) or \( a_S \) (nautical mile\(^2\)):

\[
n = \rho_a a_s, \quad \text{and} \quad n_b = \rho_a a_S.
\]

(1.20a)

(1.20b)

Values for \( \rho_a \) and \( n_b \) may be converted from units of number to kg, i.e. mass or biomass, using an appropriate factor with units of kg number\(^{-1}\) (e.g. see Hewitt and Demer, 1993; Barange et al., 1996; Demer et al., 2012).

1.2.6 Sensitivities

As previously mentioned, many of the above terms and equations used to process echosounder data originated in radar theory. Below, some of these terms are equated with those from classical sonar theory.

1.2.6.1 Transmit sensitivity

Transmit voltage sensitivity, \( S_u \) (dB re 1 \( \mu \)Pa V\(^{-1}\)) (also known as the transmit voltage response), is defined as the \( S_L \) resulting from a 1-V rms continuous-wave signal applied to a point on the electrical side of the transducer (Figure 1.4). It is expressed by substituting \( \varepsilon \) with \( (1 \text{ V})^2 \) in Equation (1.8):

\[
S_u = 10\log_{10} \left( \frac{(1 \text{ V})^2}{\varepsilon^2_{ref}} \right) = 10\log_{10} \left( \frac{(1 \text{ V})^2}{\varepsilon_{ref}} \frac{10^{-a_0/10} \rho_w c_w}{4\pi r_0^2} \right), \quad \text{or}
\]

(1.21a)

\[
S_u = \frac{(1 \text{ V})^2}{\varepsilon_{ref}} \frac{10^{-a_0/10} \rho_w c_w}{4\pi r_0^2}.
\]

(1.21b)

Rearranging Equation (1.21a):

\[
S_L = S_u + 20\log_{10} \left( \frac{\varepsilon_{ref}}{1 \text{ V}} \right).
\]

(1.21c)

Rearranging Equation (1.21b):

\[
g_0 = s_u \frac{\varepsilon_{ref}}{(1 \text{ V})^2} \frac{A\pi r_0^2}{4\pi r_0^2} 10^{a_0/10}.
\]

(1.21d)

Therefore, \( g_0 \) is a function of \( s_u \) and \( \varepsilon_{ref} \).

Transmit current sensitivity, \( S_i \) (dB re 1 \( \mu \)Pa A\(^{-1}\)) (also known as the transmit current response), is defined as the \( S_L \) resulting from a 1-A rms continuous-wave signal applied to a point on the electrical side of the transducer (Figure 1.1). It is similarly expressed by substituting \( \varepsilon \) with \( (1 \text{ A})^2 \) in Equations (1.7) and (1.8):

\[
S_i = 10\log_{10} \left( \frac{(1 \text{ A})^2}{\varepsilon_{ref}} \right) = 10\log_{10} \left( \frac{(1 \text{ A})^2}{\varepsilon_{ref}} \frac{10^{-a_0/10} \rho_w c_w}{4\pi r_0^2} \right), \quad \text{or}
\]

(1.22a)

\[
s_i = (1 \text{ A})^2 \varepsilon_{ref} \frac{10^{-a_0/10} \rho_w c_w}{4\pi r_0^2}.
\]

(1.22b)

Rearranging Equation (1.22a):
Calibration of acoustic instruments

\[ SL = S_i + 20 \log_{10} \left( \frac{u_{et}}{1 V} \right). \]  
\[ (1.22c) \]

Rearranging Equation (1.22b):

\[ g_0 = s_i \frac{1}{(1 A)^2} \frac{r}{r_0} 4 \pi r_0^2 10 \alpha r_0/10. \]  
\[ (1.22d) \]

Figure 1.4. Transmit voltage \((s_u)\) and current sensitivities \((s_i)\) are defined as the source level \((SL)\) resulting from a continuous-wave rms signal, \(u_{et} = 1 V\) or \(i_{et} = 1 A\), respectively, applied to a point on the electrical side of the transducer.

1.2.6.2 Receive sensitivity

The receive voltage sensitivity, \(M_u\) (dB re 1 V μPa⁻¹) (also known as the open-circuit receiving sensitivity), is defined as the open-circuit rms voltage at a point on the electrical side of the transducer resulting from a plane wave with \(i_{ref}\) applied to the transducer surface (Figure 1.5) (Andersen, 2006):

\[ M_u = 20 \log_{10} \left( \frac{u_{et}}{1 V} \right) - 10 \log_{10} \left( \frac{i}{i_{ref}} \right). \]  
\[ (1.23) \]
Figure 1.5. Receive voltage sensitivity \((M_u)\) is defined as the open-circuit voltage at a point on the electrical side of the transducer resulting from a plane wave with an rms pressure of 1 \(\mu\)Pa applied to the transducer surface.

When the transducer is connected to an echosounder receiver, the received voltage, \(u_{er}\) (V), is equal to the transducer voltage, \(u_{et}\) (V), electrically divided by \(z_{et}\) and the echosounder receiver electrical impedance, \(z_{er}\) (\(\Omega\)) (Figure 1.6):

\[
u_{er} = \frac{u_{et}z_{et}}{z_{er} + z_{et}}.
\]

(1.24)

In the ideal case when \(z_{er} = z_{et}\), the receiver is “matched” to the transducer and the \(p_{er}\), proportional to \(u_{er}^2\), is maximally transferred to the echosounder receiver.

\[
u_{er} = \frac{u_{et}z_{et}}{2z_{et}} = \frac{u_{et}}{2}, \text{ and}
\]

(1.25)

\[
p_{er} = \frac{u_{er}^2}{z_{et}} = \frac{u_{et}^2}{4z_{et}}.
\]

(1.26)

**Figure 1.6.** The voltage received by an echosounder (\(u_{er}\)) in response to an acoustic pressure on the transducer face is a fraction of the transducer voltage (\(u_{et}\)) defined by the echosounder receiver and transducer impedances (\(z_{er}\) and \(z_{et}\), respectively).

The \(p_{ero}\) is expressed by combining Equations (1.1), (1.3), (1.11) and (1.26):

\[
p_{ero} = i a_r, \text{ and}
\]

(1.27a)

\[
\frac{u_{et}^2}{4z_{et}} = \frac{i \lambda^2 g_0}{4\pi}.
\]

(1.27b)

Dividing both sides by \((1 V)^2\) and \(i_{ref}\):

\[
\frac{u_{et}^2}{(1 V)^2 i_{ref} 4z_{et}} = \frac{i \lambda^2 g_0}{(1 V)^2 i_{ref} 4\pi}.
\]

(1.27c)

Converting to decibels and rearranging:
20\log_{10} \left( \frac{u_{et}}{V} \right) - 10\log_{10} \left( \frac{i}{i_{ref}} \right) = 10\log_{10} \left( g_0 \frac{\lambda^2}{4\pi} i_{ref} \frac{4\pi \lambda}{V} \right) \tag{1.27d}

Substituting Equation (1.23):

\[ M_u = 10\log_{10} \left( g_0 \frac{\lambda^2}{4\pi} i_{ref} \frac{4\pi \lambda}{V} \right), \text{ or} \tag{1.27e} \]

\[ m_u = g_0 \frac{\lambda^2}{4\pi} i_{ref} \frac{4\pi \lambda}{V}. \tag{1.27f} \]

Rearranging:

\[ g_0 = \frac{4\pi (1V)^2}{\lambda^2} \frac{m_u}{i_{ref}}. \tag{1.27g} \]

which indicates that \( g_0 \) may be modulated by changes in \( m_u, \rho_w, c_w, \) and \( z_{et}, \) which may covary (Demer and Renfree, 2008).

The reciprocity parameter, \( j \) (dimensionless), is the ratio of the response of a reciprocal transducer acting as a receiver, i.e. \( m_u, \) to its response as a transmitter, i.e. \( s_i, \) (Foldy and Primakoff, 1945; Carstensen, 1947; Primakoff and Foldy, 1947; Urick, 1983). Substituting Equations (1.22b) and (1.27f), and ignoring absorption:

\[ j = \left( \frac{m_u}{s_i} \right)^{0.5} = \left( \frac{i_{ref} \frac{\lambda^2}{4\pi} (1V)^2}{\frac{4\pi \lambda}{V}} \right)^{0.5} \tag{1.28a} \]

Cancelling and rearranging:

\[ j = \frac{2k_0 p_{ref}^2}{\rho_w c_w} \left( W^{-1} \right). \tag{1.28b} \]

Substituting \( p_{ref} = 10^{-6} \text{Pa}, \) \( p_a = \text{kg m}^{-1} \text{s}^{-2}, \) and \( W = \text{kg m}^2 \text{s}^{-3} :)\]

\[ j = \frac{2k_0}{\rho_w c_w} 10^{-12} \left( \frac{\text{kg}}{\text{m}^8 \text{s}^2} \right). \tag{1.28c} \]

1.3 Common echosounder types

There are several types of echosounders that are now commonly used for fishery applications. These include single-beam, split-beam, and multibeam echosounders, and single-beam (conventional) and split-beam (interferometric or phase-measuring) sidescan sonars. Less commonly used devices include acoustic Doppler current profilers (ADCPs), acoustic cameras, and scanning sonars (see Chapter 5). The method used to calibrate an echosounder generally depends on the system’s ability to estimate the bearing angles, \( \alpha, \beta, (\^), \) to a target.

1.3.1 Single-beam

For single-beam echosounders, the bearing angles in the direction of the maximum transducer directivity pattern, conventionally \( \alpha = 0^\circ, \beta = 0^\circ \) (see Figure 1.3) may be estimated by positioning a spherical target such that its echo is maximized. Then, if the sphere \( \sigma_{bs} \) and the \( h(\alpha, \beta) \) are known for one or more colocated transducers, the off-axis elevation angles can be estimated from the differences in on-axis and off-axis
measures of sphere $\sigma_{bs}$ (Warren and Demer, 2010). For echoes from targets with unknown $\sigma_{bs}$ values, the natural convolution of the target $\sigma_{bs}$ values and $b(\alpha, \beta)$ causes ambiguity in the estimated bearing angles. However, if an ensemble of backscatter measurements is made randomly throughout the transducer beam, the $b(\alpha, \beta)$ can be deconvolved to estimate the probability density function (pdf) of the actual $\sigma_{bs}$ values (Clay and Medwin, 1977).

1.3.2 Split-beam

Split-beam echosounders can more accurately estimate bearing angles to targets within the main lobe of the transducer, provided the targets are resolvable in range and constrained to a narrow range of angles compared to the beamwidth. Split-beam transducers are separated into subarrays (e.g. four quadrants for a circular transducer). The effective separation between the subarray centres ($d_{eff}$) produces small range differences to the target that are observed as phase-differences in the received signals between each pair of quadrants (Burdic, 1991). In two dimensions and using the notation defined by Figure 1.7, this electrical phase difference, $\varphi_e$ (rad), is:

$$\varphi_e = \frac{2\pi}{\lambda} r_2 - r_1 \approx k d_{eff} \sin(\theta) = A \sin(\theta),$$

(1.29)

where $r_2$ and $r_1$ are the ranges between the centres of each subarray and the target, $\lambda$ is the acoustic wavelength at the transducer, $k$ is the acoustic wavenumber ($k = 2\pi f / c_w$), and $\theta$ is the angle to the target relative to the transducer beam axis, i.e., $\alpha$ or $\beta$. The product $kd_{eff}$ is termed angle sensitivity, $A$ (electrical°/geometric°). The $d_{eff}$ is known a priori and the sound speed, $c_w$ (m s$^{-1}$), at the transducer can be measured.

If targets are not resolvable, measurements of $\varphi_e$ may not be possible or accurate (e.g. Hewitt and Demer, 1991; Foote, 1996; Soule et al., 1996; Demer et al., 1999). Also, $\varphi_e$ may not be measurable for targets away from the beam axis if the pulses are short and the targets are extended in range, because the pulses received on each subarray may not sufficiently overlap (Lurtzon, 2000). In general, the uncertainty associated with $\varphi_e$ increases as the target occupies more of the beam (Jin and Tang, 1996) and the ratio of signal and noise intensities, $r_{sn}$ (dimensionless), decreases (Demer et al., 1999). For ideal
point targets, the standard deviation ($\sigma$) for a phase-difference measurement is (Burdic, 1991; Demer et al., 1999):

$$\sigma = \sqrt{\frac{2}{r_{sn}}} .$$

(1.30)

### 1.3.3 Sidescan

Sidescan sonars are a type of single-beam echosounder which have a transducer (stave) that is several tens of wavelengths long in one direction and a few wavelengths or less in the orthogonal direction, producing an asymmetric beam with an angular resolution that is typically $\leq 1^\circ$ or $\geq 50^\circ$, respectively. Sidescan sonars have the same target-angle ambiguity as other single-beam echosounders. However, for imaging seabed backscatter, sidescan sonars use an estimate of the sound speed along the propagation path to convert propagation delays to target ranges and, assuming the seabed is flat and horizontal, or known from an independent source, to bearing angles.

There are also split-beam (interferometric) sidescan sonars that are constructed by stacking two or more single-beam transducers (Figure 1.8). The backscatter from a target constrained to a narrow range of the beamwidth is manifested as a split-beam phase difference that is used to estimate the bearing angle (see Section 1.3.2). Developed for seabed mapping, the bearing angles are only estimated in one plane (typically athwartships). The bearing angles in the orthogonal plane (typically alongships) are estimated as the direction of the beam axis (see Section 1.3.1).

![Figure 1.8. The geometry for bearing estimations using split-beam sidescan sonar. Two single-beam sidescan transducers, separated by a distance $d_{eff}$ ($d$ in the figure), are used to measure electrical phase differences and thus estimate the bearing angles to the seabed.](image)

### 1.3.4 Multibeam

Multibeam echosounders are constructed from an array of transducer elements. In the common Mills cross configuration (Figure 1.9), for example, a line array (similar to a side-scan transducer) is used to transmit a fan beam, and an orthogonal array of elements is used to electronically form multiple, typically hundreds, of perpendicular receive beams. This combination of transmit and receive beams creates a swath of narrow, typically $<1^\circ$, “pencil” beams.
Multibeam transducers can also be constructed using two- or three-dimensional arrays of transducer elements arranged in a plane, cylinder, or other geometry. To transmit multiple beams, distinct frequencies or orthogonal phase-codes (unambiguous signals) are used for each beam to avoid cross-talk between beams. The number of transmit beams and their signal levels are limited by the power densities that can be achieved without non-linear effects (Tichy et al., 2003; Korneliussen et al., 2008) or damage to the transducer elements. To simultaneously steer and adjust the directivities of multiple receiver beams, the digitized signals from the transducer elements are typically “beam-formed” using complex weights, time delays, or both.

In multibeam echosounders, target-bearing angles are estimated in different ways depending on the nature of the target. For targets that span a small portion of the beam-width, the beamforming may include subarrays and split-beam processing. For targets that span a large portion of the beamwidth, the beamformer output is largest at the most probable bearing angle. Although extended to multiple beams, this filtering is analogous to bearing angle estimation in a single-beam echosounder.

### 1.3.5 Wide-bandwidth

Wide-bandwidth echosounders transmit signals spanning several octaves (Stanton et al., 2010), and the backscatter vs. frequency (spectra) is often used to classify the acoustic targets (Woodd-Walker et al., 2003; Imaizumi et al., 2008; Johnsen et al., 2009). For example, low-frequency backscatter spectra from fish may indicate mechanical resonances in their swimbladders (Hersey et al., 1962), midfrequency spectra may differentiate backscatter from fish and zooplankton (e.g., Demer et al., 2009b), and high-frequency spectra may differentiate backscatter from marine organism and turbulent microstructure (e.g., Lavery et al., 2010a). Because the transceiver electronics, transducer characteristics, and sound propagation medium all affect the frequency spectra, it is necessary to quantify and deconvolve these effects from the target’s backscattering spectra. For example, most acoustic transducers have frequency-dependent beam-widths and sensitivities (Rogers and Van Buren, 1978), which must be taken into account when calibrating a wide-bandwidth echosounder (Islas-Cital et al., 2010).

### 1.3.6 Scanning

Scanning sonars electrically or mechanically steer acoustic beams to insonify and image scatterers throughout a large swath, potentially 360° horizontally, or hemisphere. The effective range of a scanning sonar depends on the signal type and bandwidth, and
the scatterers. Scanning sonars have long been used by fishers to detect fish schools. They are used increasingly to observe fish behaviour and quantify their aggregation biomass.

### 1.3.7 Acoustic Doppler

An acoustic Doppler current profiler (ADCP) is an echosounder that measures relative water speed inferred from the shift in frequency content of the received vs. transmitted pulses. ADCPs typically have three or four narrow-beam transducers oriented in different directions to measure water velocity (speed and direction). The frequency shift is proportional to the velocity of the target relative to the ADCP beam axis. ADCP measurements are calculated with the assumption that the water speed is equivalent to that of the particulates or organisms contributing to the backscatter. ADCPs may also use echoes from the seabed to estimate movement of the instrument platform relative to earth. In addition to measuring current and platform velocities, most ADCPs measure echo intensity. However, the quality of echo intensity data from ADCPs is generally limited by a low dynamic range, temperature-sensitive gain, and the processing of wide-bandwidth signals. Because ADCPs are available on many research vessels, their data is often used opportunistically for various types of surveys, e.g. biomass or suspended sediment.

### 1.3.8 Imaging

An underwater acoustic camera uses echo intensity and propagation-delay data from an array of synchronous echosounder beams to create a two- or three-dimensional image of scatterers within the field-of-view. Some acoustic cameras use beamforming to amplify the signal in certain directions while suppressing the others. This method is fast and robust, but the beam sidelobes can produce ghost images. Alternatively, some acoustic cameras use acoustic lenses to create the receiver array. Acoustic cameras are used increasingly to observe animals in shallow and turbid water, and near the seabed.

### 1.4 Common transducer platforms

Echosounders and their transducers are mounted on a variety of platforms ranging from mobile research vessels and autonomous vehicles to relatively stationary landers and buoys. In each case, echosounder calibrations are generally performed using the sphere method (see Chapter 2).

#### 1.4.1 Vessel

An acoustic system is generally installed on a vessel when its primary purpose is to survey targets spanning large areas, map their distributions, and estimate their abundance, e.g. using echo integration methods (Dragesund and Olson, 1965). On a vessel, transducers are mounted either on the hull or keel or over the side using a pole or towed-body. In most cases, the beam axis of each transducer is oriented downward, the cables are routed to minimize electrical interference and mechanical vibration, and calibration equipment is available to facilitate routine calibrations.

#### 1.4.1.1 Hull

Transducers may be permanently mounted on a vessel hull or keel or on a small pod extending a short distance below the hull (Simmonds and MacLennan, 2005), optimally below any aerated water to provide low-noise operation. The main disadvantages of hull-mounted installations are cost and the signal loss due to motion-induced bubbles forced along the ship’s hull in adverse weather conditions (Carstensen, 1947; Dalen
and Lovik, 1981; Shabangu et al., 2014). Hull mounting often requires architecture and engineering, dry docking, fabrication, installation, and testing. Therefore, it is done when the benefit of dedicating equipment to a vessel outweighs the installation costs.

1.4.1.2 Centreboard

It is now common to mount transducers on the bottom surface of a retractable centreboard. The centreboard positions the transducers at variable, user-defined depths below the hull, typically 5–10 m depth, to reduce vessel motion and noise from bubbles.

1.4.1.3 Pole

Transducers may be mounted on the end of a pole affixed to the side, bow, or stern of a vessel (Figure 1.10). Usually, the pole length positions the transducers a few metres below the water surface close to the depth of the vessel hull. The transducer depth may be fixed or adjustable. Transducers deployed deeply on a pole have advantages similar to centreboard-mounted transducers. However, poles are smaller than centreboards and can be retrofitted to small and large vessels (Gangl and Whaley, 2004; Cutter and Demer, 2007). Poles should be rigid during surveys and retractable for transiting and docking.

Figure 1.10. Pole-mounted transducers. Hydraulically retractable poles (left) are used to mount multibeam (starboard) and multifrequency split-beam transducers (port) on a small craft. A manually retractable pole is used aboard a fishing boat and other platforms of opportunity (right).

1.4.1.4 Towed and cast

Towed bodies and cast platforms may be used to deploy transducers alongside, behind, or beneath the vessel (Figure 1.11). They are portable and may be moved regularly among vessels that are not equipped with permanently mounted scientific echo-sounders, such as chartered commercial vessels. During surveys, towed bodies are often positioned 4–6 m below the surface to avoid bubble noise and side-lobe echoes from the hull. However, when surveying fish or measuring backscattering cross sections at large depths, towed bodies or cast platforms may position the transducers 10s–100s of metres deep. Deep deployments reduce the ranges between the transducers and the acoustic targets, improving the spatial resolution of the measurements (Ona and Mitson, 1996).
1.4.2 Autonomous moving

Acoustic transducers may be deployed on autonomous underwater vehicles (AUVs), gliders, drifters, buoys, and landers. Such platforms may be used to collect data independently from a surface vessel and over large distances (e.g. several 1000 km), long time-scales (e.g. months), or both.

1.4.2.1 Propelled

AUVs are self-propelled and perform preprogrammed underwater missions (e.g. Fernandes et al., 2003). They are often torpedo shaped, 2–10 m long, and 0.2–1.3 m in diameter (Figure 1.12). Echosounder transducers may be mounted on AUVs in down-, up- or side-projecting orientations (Scalabrin et al., 2009).

Underwater gliders are autonomous vehicles that undulate along transects using wings for lift and low-power electro-mechanics to control pitch, roll, and buoyancy (Rudnick et al., 2004). They may survey several thousand kilometres over several months (Eriksen et al., 2001; Davis et al., 2002). Sensors require low power and low hydrodynamic resistance (drag).
1.4.2.2 Drifting
Transducers mounted on drifters can be used to investigate organisms or processes that are dependent on oceanic currents. Once deployed, a Lagrangian drifter moves with water currents at the sea surface or at a specific water depth or density, depending on buoyancy. Some drifters can be programmed to change buoyancy and, therefore, may reside at user defined water depths throughout their deployment.

1.4.1 Autonomous stationary
Transducers may be mounted in a fixed location to observe the physical and biological environment vs. time. Stationary platforms facilitate observations on many temporal scales and do not require an operator or propulsion. Echosounder power may be supplied via cable from shore, or locally by batteries, perhaps recharged by photovoltaic cells or a wind generator.

1.4.1.1 Buoy
An echosounder system on a moored buoy may be used to monitor a selected location (Brierley et al., 2006). Often, the echosounder transceiver, GPS receiver, computer, batteries, radio, and transducer are contained in the buoy float. The transducers, oriented horizontally or vertically, are positioned multiple metres below the sea surface to reduce bubble noise and the potential for biofouling (Figure 1.13).

Figure 1.13. Buoys fitted with dual-frequency and split-beam echosounders, and acoustic Doppler current profilers.

1.4.1.2 Lander
Landers are stationary platforms positioned on or near the seabed. Transducers mounted on landers usually project up or to the side. Landers are often used to make measurements spanning from near the seabed to the sea surface over long periods of time. They may also be used in areas where buoys are adversely subjected to weather or vandalism.

1.5 Calibration methods
Acoustic instrument calibrations estimate the accuracy (systematic error) and precision (random error) in the transduction processes and measurements of the resulting electrical signals. Each signal-processing step, from generation to digitization, may affect
Calibration of acoustic instruments

the system calibration. This does not necessarily mean that each process must be calibrated independently; it is generally more accurate and efficient to calibrate the whole echosounder system. In any case, acoustic instrument calibrations should be performed over the range of environmental conditions, particularly water temperature, $T_w$ (°C), and pressure, $P_w$ (mbar), or depth (Leroy and Parthiot, 1998) encountered during the measurements made using the instrument.

There are a number of methods to calibrate acoustic instruments (Urick, 1983; Bobber, 1988; Simmonds and MacLennan, 2005). These methods may be categorized as component or system calibrations.

1.5.1 Component

A component calibration requires identification and performance evaluation of each system component. It requires an assumption that the systematic errors from each component are independent and linearly sum to indicate the total systematic error (bias) in the measurement system. A component calibration may be useful to identify the major sources of systematic error and to confirm that the performance of each component is within the specification tolerance. However, a component calibration generally requires additional calibration instrumentation, expertise, and time, with its result affected by the accumulation of errors. Described below are some methods for characterizing some echosounder components, principally the transducer.

1.5.1.1 Transducer reciprocity

The sensitivities of a reciprocal transducer, i.e. one in which the energy losses on transmission are the same as on reception, may be measured by a reciprocity calibration (Schottky, 1926; Foldy and Primakoff, 1945). This involves three uncalibrated transducers including a projector, hydrophone, and the transducer to be calibrated (MacLennan and Simmonds, 1992). The three transducers are positioned in a triangle with distances between them being greater than their far-field ranges (Urick, 1983). For each set of measurements, the acoustic axes of each transducer pair must be aligned. For further details, see Urick (1983) and Bobber (1988). Reciprocity calibrations may be practical for calibrating echosounder transducers in tanks, but probably not in the field.

1.5.1.2 Transducer impedance

Echosounder transducers may be calibrated (MacLean, 1940; Sabin, 1956) and their performance monitored by measuring transducer impedance (Demer and Renfree, 2008). Transducer impedance and, therefore, transducer performance may change for a variety of reasons, e.g. due to changes in water temperature and pressure, electrical isolation, or damage to the piezoelectric elements. Impedance measurements may be made with an impedance analyser. However, some instruments have integrated tests for rapidly monitoring change in the transducer impedance.

1.5.1.3 Element analysis

Transducers have transmit- and receive-beam directivities that depend on the ratios of their sizes and the acoustic wavelengths. To increase the beam directivity and reduce its sidelobe intensities, and perhaps steer its axis, echosounder transducers are often composed of multiple elements that are electrically and spatially weighted, and beam-formed, respectively (Wilson, 1988).

Some echosounder systems provide raw data from each transducer element. Therefore, theoretically, it is possible to characterize the performance of each transducer element and use those results to calculate the performance of the transducer for a variety of
transmit and receive configurations. Practically, however, this strategy could require the characterization of numerous transducer elements and many assumptions about manufacturing tolerances; the resulting estimate of transducer performance would be subject to the accumulation of estimation errors.

More convenient are tools, available in many echosounders, that monitor the performances of individual elements and their transceiver channels during transmission and reception (e.g. Figure 1.14). This can be helpful as the beamforming performance of a numerous-element transducer array may not be obviously degraded by a few defective elements or transceiver channels.

![Figure 1.14](image)

Figure 1.14. The ME70 BITE utility (Simrad, 2012) displays the average transmit level for each transducer element and transceiver channel. In this example, one element (blue) has a weak transmit level compared to surrounding elements. Elements associated with noisy channels on a transceiver (black) are disabled.

If one or more transducer elements or transceiver channels has become defective, the directivity pattern resulting from the functioning elements may be simulated to learn if the system performance has likely changed and, if so, what corrections may be applied or what maintenance actions are required (Figure 1.15).
Figure 1.15. Beamforming simulation with and without the disabled elements shown in Figure 1.14. In this example, the sidelobe levels (left plot) have increased substantially (>6 dB), and exceed the specification. However, the loss of signal level in the main lobe (right plot) is small (~−0.14 dB), so the change in system performance may be negligible, depending on the application.

1.5.2 System

A system calibration evaluates the collective performance of the system components, such as the transmit and receive electronics, connectors, cables, and transducers. A system calibration can be performed following the recommended sphere method (Foote et al., 1987), detailed in Chapter 2. When appropriate, other methods, briefly described below, can be used.

1.5.2.1 Self-reciprocity

A self-reciprocity calibration (Widener, 1980a, 1980b; Urick, 1983; Bobber, 1988) is analogous to a sphere calibration (Foote et al., 1987; Sawada and Furusawa, 1993; Vagle et al., 1996) when the echo from the target is replaced by the pulse transmitted from a virtual transducer (see Section 1.2.5.1). To do this, the transducer beam is accurately and precisely aligned perpendicularly to a mirror surface, e.g. a flat air–water interface that provides near-total reflection. Alignment is critical. Then, the received echo apparently originates from a virtual transducer that is twice the range between the real transducer and the reflecting surface (Carstensen, 1947; Urick, 1983; Bobber, 1988). Self-reciprocity calibrations may be performed in tanks, but they may not be practical for field calibrations because hull-mounted transducers are generally not oriented towards the sea surface.

The accuracy of a self-reciprocity calibration depends on the near-total reflection which may be degraded by misalignment and the roughness of the surface relative to the acoustic wavelength (Patterson, 1967). If the beam direction varies randomly about the perpendicular orientation, then a number of measurements may be averaged to obtain a precise, if inaccurate, calibration.

To grossly assess and compare echosounder performance, the seabed may be used as the reflecting surface (e.g. Stewart et al., 1994). However, compared to an air–water interface, seabed reflectivity is lower and more unpredictable and unstable. This is because there are a multitude of substrate types; all have a lower impedance contrast relative to air and many have dynamic scattering properties due to physical (e.g. waves and currents) and biological (e.g. burrowing animals and shellfish) processes.

1.5.2.2 Reference

A reference transducer has well characterized (usually by the manufacturer) transmit response, receive sensitivity, and beam directivity vs. frequency. To calibrate an echosounder, the reference transducer is accurately and precisely positioned on the beam axis in the far-field of the echosounder transducer (see Section 2.1.2.4). The echosounder transmit response is calibrated when the reference transducer receives a transmitted signal. The echosounder receive sensitivity is calibrated when the reference transducer transmits a signal of known intensity. This method may be used to calibrate echosounders in a tank, but the uncertainties associated with positioning and aligning the reference transducer in the field make this method impractical for many studies (MacLennan and Simmonds, 1992; Hwang et al., 2007).

Reference transducers are sometimes used to measure transmit pulse shape and source level, and to transmit and receive frequency responses. These parameters are relatively insensitive to the position of the reference transducer.
1.5.2.3 Comparison

The performance of an uncalibrated echosounder may be evaluated by comparing it to the performance of a calibrated echosounder. If the echosounders in the comparison are mounted on different platforms (e.g. Foote et al., 1987; Fernandes et al., 2000; De Robertis and Wilson, 2006; De Robertis et al., 2008), then the comparisons may be affected by any differences in the echosounder specifications and settings (Jech et al., 2005), the temporal and spatial distributions and behaviours of the target species, the temporal and spatial characteristics of the sampling, the radiated vessel noise (Mitson and Knudsen, 2003; De Robertis et al., 2008), and the weather. Measurements from each system may be statistically compared with a linear regression that accounts for the systematic and random errors in both sets of measurements. If the values from the different systems are statistically equivalent, then both systems may be calibrated to the same tolerances. If the measurements are statistically different, then other methods are required to understand the causes of the differences. Similarly, measurements made with uncalibrated echosounders have been compared to independent measures, e.g. net samples (see Section 5.5).
2 Sphere calibration

2.1 Introduction

A sphere calibration measures the overall performance of an echosounder using the reflection from a solid sphere of known backscattering cross section, $\sigma_{bs}$ (m$^2$), also known as target strength (TS; dB re 1 m$^2$). The result is used to adjust the echosounder gain, $g_0$ (dimensionless), and the filter attenuation correction factor, $s_{a_{corr}}$ (dimensionless), to optimize the accuracy of acoustic measurements of $\sigma_{bs}$ and volume backscattering coefficient, $s_v$ (m$^2$ m$^{-3}$), also known as volume backscattering strength, $S_v$ (dB re 1 m$^2$ m$^{-3}$). In addition to providing these estimates, a conventional sphere calibration also provides a check of the beam-pattern characteristics. The details of the calibration procedure vary with the echosounder type, e.g. single-beam or split-beam, and transducer platform, e.g. vessel or towed body, but the aim and general procedure are the same.

Basically, a sphere of known $\sigma_{bs}$ is placed into the beam of the acoustic transducer, and the echo received by the echosounder is recorded. The sphere is moved throughout the acoustic beam to measure its variation in sensitivity due to the beam directivity, $d(\alpha, \beta)$ (dimensionless). It is especially important to obtain data when the sphere is located in the centre of the acoustic beam (i.e. the axis of maximum sensitivity) where the sphere echo is strongest, as these measurements are used for estimating $g_0$ and $s_{a_{corr}}$.

Sphere calibrations should be conducted often and over the range of water temperature, $t_w$ (°C), salinity, $s_w$ (psu), sound speed, $c_w$ (m s$^{-1}$) and pressure, $p_w$ (dbar), that the transducers will encounter during the measurements. At a minimum, calibrations should be performed before and after the acoustic instrument is used to make measurements.

The remainder of this section describes the equipment, procedures, and analysis for conducting calibrations of acoustic equipment using spheres. Chapter 3 explores the uncertainty in sphere calibrations. Chapter 4 provides additional details that are specific to some commonly used acoustic equipment.

2.1.1 Equipment

A sphere calibration requires some specialized equipment including a sphere, apparatus to suspend and move the sphere within the transducer beam, and sensors to measure $t_w$, $s_w$, $c_w$, and $p_w$ at the transducer, at the sphere, and between them. Software is also needed to calculate estimates of $g_0$ and $s_{a_{corr}}$ from the sphere-$\sigma_{bs}$ measurements.

2.1.1.1 Sphere properties

A calibration sphere should have a known and stable $\sigma_{bs}$ (MacLennan, 1981; Foote, 1982, 1983b). They are usually made from tungsten carbide with 6% cobalt (WC) or ≥99.99% electrical-grade copper (Cu). Cu spheres are machined to the desired sizes. WC spheres are manufactured using a powder-metallurgy and polishing process.

Generally, calibration spheres do not have traceable origins (i.e. materials and manufacturing) or documented metrological estimates of their $\sigma_{bs}$ (i.e. measurements using national or international standards). Nevertheless, the precise composition of a calibration sphere is critical and should be verified since its $\sigma_{bs}$ varies with the material properties (MacLennan, 1982; Foote and MacLennan, 1984). For example, the density of a WC sphere depends on the amount of cobalt binder. Calibration spheres should be weighed and their calculated densities compared to the values in Table 2.1.
The $\sigma_{bs}$ of the sphere is calculated using a theoretical model for the reflection of sound by elastic spheres (MacLennan, 1981). The model requires estimates of the sphere size and its material properties (Foote and MacLennan, 1984; MacLennan and Dunn, 1984), the acoustic frequency, $f$ (Hz), and the density, $\rho_w$ (kg m$^{-3}$), and $c_w$ of the surrounding water. A web-based application is available to calculate $\sigma_{bs}$ using the model in MacLennan (1981) (http://swfscdata.nmfs.noaa.gov/AST/SphereTS/).

Table 2.1. Material properties for tungsten carbide with 6% cobalt binder (WC) and copper (Cu), from *Foote and MacLennan (1984) and MacLennan and Dunn (1984).* It is highly recommended to weigh spheres and validate their densities.

<table>
<thead>
<tr>
<th>Property</th>
<th>Cu$^1$</th>
<th>WC$^1$</th>
<th>WC$^2$</th>
</tr>
</thead>
<tbody>
<tr>
<td>Density (kg m$^{-3}$)</td>
<td>8.947</td>
<td>14.900</td>
<td>14.900</td>
</tr>
<tr>
<td>Longitudinal sound speed (m s$^{-1}$)</td>
<td>4.760</td>
<td>6.864</td>
<td>6.853±19</td>
</tr>
<tr>
<td>Transverse sound speed (m s$^{-1}$)</td>
<td>2.288.5</td>
<td>4.161.2</td>
<td>4.171±7</td>
</tr>
</tbody>
</table>

Sphere diameters are chosen to have stable and strong reflections throughout the echosounder frequency-bandwidth, and the expected water properties during calibration (Foote, 1982). The sphere-$\sigma_{bs}$ can have large and abrupt changes in magnitude over small frequency ranges (MacLennan, 1981; Sun *et al.*, 1991) (Figure 2.1). Spheres with such regions close to the operating frequency of the echosounder should be avoided. Therefore, one or more different spheres may be needed to calibrate echosounders operating at different frequencies (e.g. Foote, 1990).

Use the fewest number of spheres necessary to accomplish the calibration. Deploying multiple spheres sequentially takes time and, during the transitions, the suspension lines may become caught or tangled. Deploying multiple spheres simultaneously, one above the other, can save time, but may increase calibration error due to forward scattering (MacLennan, 2011).

Commonly used sphere sizes and materials are given in Table 2.2. These are for narrow-band systems and have been chosen such that their $\sigma_{bs}$ values are insensitive to small changes in $f$, $t_w$, and acoustic pulse duration, $\tau$ (s) (Table 2.3) (Foote, 1982, 2007; Miyanohana *et al.*, 1993).
Table 2.2. Recommended spheres for calibrating common echosounder frequencies. The spheres are made from copper (Cu) or tungsten carbide with 6% cobalt binder (WC).

<table>
<thead>
<tr>
<th>Frequency (kHz)</th>
<th>Material/diameter (mm)</th>
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<tbody>
<tr>
<td>12</td>
<td>Cu/45.0</td>
</tr>
<tr>
<td>18</td>
<td>WC/38.1, Cu/64.0, Cu/63.0</td>
</tr>
<tr>
<td>27</td>
<td>Cu/42.0</td>
</tr>
<tr>
<td>38</td>
<td>WC/38.1, Cu/60.0</td>
</tr>
<tr>
<td>50</td>
<td>Cu/45.0</td>
</tr>
<tr>
<td>70</td>
<td>WC/38.1, Cu/32.0</td>
</tr>
<tr>
<td>120</td>
<td>WC/38.1, Cu/23.0</td>
</tr>
<tr>
<td>200</td>
<td>WC/38.1, Cu/13.7</td>
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<tr>
<td>333</td>
<td>WC/22.0</td>
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</tbody>
</table>

Calibration spheres should have $\sigma_{bs}$ values (see Table 2.3) which provide a signal-to-noise ratio, $r_{sn}$ (dimensionless), that is $\geq 100$ (20 dB) at the calibration location. The spheres should not have nicks, scratches, dents, or corrosion. They should be cleaned and dried after each use and stored in a protective container. Spheres are occasionally lost overboard during calibrations, so it is prudent to have spare spheres available.

Table 2.3. Approximate theoretical target strength, $T_S$ (dB re 1 m$^2$ at $r_g = 1$ m), of common calibration spheres with various diameters (mm), made from tungsten carbide with 6% cobalt (WC), and copper (Cu), at $t_w = 13.5$ °C, $s_w = 33.3$ psu, $p_w = 25.0$ dbar, $c_w = 15.00$ m·s$^{-1}$, and $\rho_w = 1025.0$ kg·m$^{-3}$. Green indicates there are no nulls within or near the signal bandwidth $b_f \approx f \pm 0.5(1/r)$, where $f$ is frequency (Hz) and $r$ is pulse duration (s). Yellow indicates a null close to the $b_f$, and red indicates a null within the $b_f$.

<table>
<thead>
<tr>
<th>Material</th>
<th>Diameter (mm)</th>
<th>$f \cdot 10^{-3}$ (Hz)</th>
<th>64</th>
<th>128</th>
<th>256</th>
<th>512</th>
<th>1024</th>
<th>2048</th>
<th>4096</th>
<th>8 192</th>
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<td>WC</td>
<td>20.0</td>
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### 2.1.1.2 Sphere suspension

Suspend the sphere by one or more monofilament lines. The line may be attached to the sphere by either gluing a loop of line into a small hole, or constructing a net bag (Figure 2.2). Holes may be drilled into Cu spheres, but must be spark-eroded into WC spheres as they are extremely hard. In either material, the hole should be small (e.g. 2-mm diameter by 3-mm deep) such that it has negligible effect on the sphere \( \sigma_{bs} \) (e.g. Foote, 1982).

![Knot Detail](image)

**Figure 2.2.** A method for tying a net around a sphere. Proceed from step (a) to (g). Tie the knots in numbered order. The interknot distance should equal one-fourth of the sphere circumference. This graphic is a revision of Figure 4 in Foote et al. (1987).

The sphere suspension method depends on the transducer mount and platform. Generally, the sphere is suspended from three monofilament lines run from reels and outrigger poles mounted on the railings of the ship (see Figure 2.3). Outriggers are used
to keep the monofilament lines from rubbing on the hull and for moving the sphere during the calibration. The line lengths are adjusted to precisely move the sphere within the acoustic beam.

Outriggers may be manual or motorized fishing outriggers with line-length indicators. Alternatively, outriggers may be constructed with manual or electric fishing reels, each attached to the inboard end of a rigid rod and the monofilament line fair-led through a ring or pulley on its outboard end. Motorized outriggers may be controlled remotely and by computer. In lieu of a mechanical or electronic line-length indicator, the line may be marked incrementally with colour.

Fill each outrigger reel with mostly non-elastic line (e.g. Dyneema, Spectra, or stainless steel). To each line, attach (i) a small swivel, (ii) monofilament line attached to the sphere, and (iii) monofilament line attached to a stabilizing weight (see Figure 2.3). To reduce uncertainty in the sphere-echo intensity and position caused by phase noise from the monofilament attachments, the three monofilament lines should be attached to three different points on the net bag. If echoes from the lines are not a concern, e.g. when calibrating lower-frequency echosounders, then loops in the monofilament lines extending from each downrigger may be joined to the loop on the sphere and a loop in a monofilament extension line attached to the weight. In any case, echoes from the swivels, sphere, and weight must be acoustically resolvable (separated). Thus, the vertical distances between the swivels, sphere, and weight should be $\geq c_w r$, e.g., $\geq (1500 \text{ m s}^{-1})(0.001 \text{ s}) = 1.5 \text{ m}$.

Figure 2.3. The three-line method for suspending a sphere below a hull-mounted transducer. The monofilament lines should be attached to the mesh bag at different locations to minimize phase noise at high frequencies (i.e., $>100$ kHz). Note, if the weight is too heavy, the net bag may break and the sphere could be lost.

Forward scatter from the swivels, lines, and knots must be negligible or else the sphere-echo intensity and phase will be degraded, especially at high frequencies (MacLennan,
To minimize forward scatter, the swivels and knots should be small and the monofilament line thin, yet strong enough to withstand the deployment, movement, and retrieval of the sphere and weight. Also, to shed bubbles, dip the sphere, swivels, knots, and lines in soapy water before deploying.

An alternative to the three-line method is to run a single monofilament line under the vessel from one side to the other. Suspend the sphere from the midpoint of the line. Adjust the sphere athwartships by releasing line on one side of the ship and retrieving an equal amount of line on the other side. Adjust the sphere alongships by moving both ends of the line forward or aft.

2.1.2 Calibration procedure

This section provides guidance on the general sphere-calibration procedure. Subsequent sections in this chapter detail each operation.

1) Stabilize the transducer, e.g. by positioning the vessel in a sheltered area (see Section 2.1.2.1).
2) Measure transducer impedance, the receiver test amplitude, or both (see Section 2.1.2.2).
3) Choose an appropriate sphere (see Tables 2.2 and 2.3).
4) Suspend the sphere in the transducer beam using outriggers and monofilament (see Sections 2.1.2.3 and 2.1.2.4).
5) Measure the $t_w$, $s_w$, and $c_w$ at the transducer, sphere, and between them (see Section 2.1.2.5).
6) Position the sphere in the centre of the beam, then move it throughout the beam (see Section 2.1.2.6).
7) Analyse the calibration data and results (see Section 2.1.2.7).
8) Retrieve the sphere and stow the gear (see Section 2.1.2.8).

2.1.2.1 Calibration location

As much as possible, calibration experiments should be conducted over the range of environmental conditions, i.e. $t_w$, $s_w$, $c_w$, and $p_w$, encountered during the survey measurements (Demer and Renfree, 2008). Calibrations should be performed in areas where the water is well mixed and relatively void of biological scatterers. The experiment should be scheduled for slack tide. The water depth should be sufficient to place the sphere in the far-field of the transducers (see Section 2.1.2.4), accounting for tides.

Depending on the location, traffic, wind, swell, and current, the sphere calibration may be conducted while the vessel drifts in the open ocean or anchored from one or more points. If the wind speed is $< -15$ knots and the swell is $< -2$ m, drifting may be most convenient. More often, however, anchoring in a sheltered bay or fjord is a better option. For portable echosounders, a wharf or large tank of water may be more convenient.

If the vessel is drifting or anchored from a single point, usually the bow, vessel, and tethered sphere tend to move in unison with any current. However, if the wind and current are from different directions, the vessel may be anchored from multiple points, e.g. the bow and stern, to keep it from swinging.

2.1.2.2 Transducer impedance

Prior to each sphere calibration, measure the transducer impedance and compare it to the manufacturer’s specifications and any previous measurements. The electrical im-
Calibration of acoustic instruments

The measurement of transducer impedance is most conveniently made using an electrical impedance meter, but can also be accomplished using a signal generator, oscilloscope, and inductive current probe (see Demer and Renfree, 2008). Regular measurements of impedance provide a useful diagnostic tool to help explain large changes in echosounder calibrations. Large changes in transducer impedance may indicate a faulty transducer and avoid a time-consuming and ultimately unsuccessful calibration. Abrupt changes in impedance may result from damage to the transducer or associated cabling, or bio-fouling on the transducer face. Some echosounders have built-in functions to monitor transducer impedance.

2.1.2.3 Sphere suspension

This section describes the three-outrigger method for sphere suspension beneath a large vessel (Figure 2.3). It can be readily adapted for other situations.

1) Locate the transducer(s) using hull drawings, sketches, or photographs. In some cases, divers may be needed to locate the transducers and to help initially position the sphere beneath them.

2) Attach a weight to the middle of a messenger line that is at least threefold the vessel’s beam. With the ends of this line affixed to each side of the bow, throw the weight forward and clear of the vessel. As the line sinks, untie the line and move the two ends aft until it is aligned with the transducer(s) to be calibrated. If needed, a skiff may be used to assist with this step.

3) If appropriate, anchor the vessel.

4) Attach three outrigger poles to the ship in positions approximating an equilateral triangle centred on the transducer(s). At the fore-aft position of the transducer (where the line is now located), place one outrigger on one side of the ship (e.g. starboard in the following example). If the transducer(s) are not centred athwartships, place the single downrigger on the side farthest from the transducers. On the other side of the ship (e.g. port in this example), place each of the other outriggers approximately equal distances fore (port-forward) and aft (port-aft) of the transducer. For transducers centred athwartships, these distances may be ~0.58 times the vessel width.

5) Attach the starboard outrigger line to the messenger line.

6) While maintaining slack in the line so that it does not catch on the hull, release line from the starboard outrigger and retrieve it on the port side. Continue until the starboard outrigger line is secured on the port side. Detach and stow the messenger line.

7) Bring the port-aft line to meet the port-forward line.

8) Bring all three lines together and attach the sphere and stabilization weight (see Figure 2.3).

9) Immerse the calibration sphere and monofilament line in a mixture of roughly 2/3 water and 1/3 liquid dishwashing soap to shed bubbles from the sphere and knots.

10) Carefully lower the weight and sphere over the side of the vessel and into the water without contacting the hull.

11) To initially position the sphere, release the same amount of line on all three outriggers. Port-forward and port-aft lines should be released at the same time to
ensure that the calibration sphere does not make contact with the hull. Lines running against the hull may become caught, abraded, or cut on protrusions such as anode blocks, bilge keels, or barnacles. If a line is caught, try to release it by moving the line forward or aft. If it cannot be released, it may be necessary to cut the caught line, retrieve the sphere with the other lines, and start again, perhaps using longer outrigger poles.

12) Let out line incrementally until the sphere is visible at a suitable range on the echosounder display. Avoid letting the sphere, or weight if used, hit the seabed. When the sphere is in the transducer beam, its echo should be received at a constant range (see Figure 2.4).

Figure 2.4. An echogram during a sphere calibration of a 38-kHz echosounder. The vertical axis is depth below the transducer (m), and the horizontal axis is time (hh:mm:ss). The red band at 0 m is the transmit pulse. Echoes are from organisms (~10–15 m), swivels (~21 m), a 38.1-mm diameter WC sphere (~26 m), a 60-mm diameter Cu sphere (~30 m), a stabilizing weight (~35 m), and the seabed (~39 m). Forward scatter from organisms, swivels, monofilament knots, and even other spheres can degrade the measurements of sphere position and the calibration accuracy, particularly at higher frequencies (MacLennan, 2011).

2.1.2.4 Sphere range

At short ranges, compensation of the sphere echo for spreading loss can vary significantly with small changes in the estimated range. Therefore, measurements of sphere \( \sigma_{bs} \) (or any other quantitatively sampled backscatter) should not be made at short ranges. Simmonds and MacLennan (2005) recommend that the range to the sphere is at least twofold the near-field range, \( r_{nf} \) (m). For a circular piston projector:

\[
 r_{nf} = \frac{d_t^2}{\lambda}, 
\]

where \( d_t \) (m) is the diameter of the transducer and \( \lambda \) is the acoustic wavelength (m). Medwin and Clay (1998) recommend that the range to the sphere is at least threefold the \( r_{nf} \):
where \( d_t \) (m) is the largest distance across the active elements in a circular piston projector. The recommendation by Medwin and Clay (1998) is 0.356-fold more conservative than that by Simmonds and MacLennan (2005).

The \(-3\,\text{dB}\) points on the main lobe of a transducer directivity pattern occur at off-axis angle \( \theta_{-3\,\text{dB}}/2 \equiv \sin^{-1}(3.2/kd_t) \), where \( k = 2\pi/\lambda \) is the wave number (Sherman and Butler, 2007). Therefore, \( d_t \) can be estimated from the beamwidth, \( \theta_{-3\,\text{dB}} \), using:

\[
d_t \approx \frac{3.2}{k\sin(\theta_{-3\,\text{dB}}/2)}.
\]

Combining Equations (2.1b) and (2.2), the recommended minimum range for the sphere (Medwin and Clay, 1998) is estimated for a selection of frequencies and transducer beamwidths (Figure 2.5).

Just as the sphere must be sufficiently outside of the transducer’s near-field, the transducer must be sufficiently outside of the sphere’s near-field. To estimate \( 3r_{nf} \) for the sphere, substitute the sphere diameter for the transducer diameter in Equation (2.1b). Furthermore, the sphere must approximate a point target, so the beam diameter must be large compared to the sphere diameter at the measurement range. For example, if \( \theta_{-3\,\text{dB}} = 7^\circ \), \( r = 15 \, \text{m} \), and the sphere diameter = 0.0381 m, then the sphere diameter is
1/48\textsuperscript{th} or 2.1\% of the beam diameter. This requirement may be important when calibrating multibeam sonars with small values of $\theta_{-3\,\text{dB}}$.

Even if the near-field range criteria are met for both the transducer and the sphere, sphere echoes can cause saturation in some echosounder receivers, particularly if they are designed for measuring long-range or small $\sigma_{bs}$ targets. When this happens, the received sphere-echo power remains constant vs. range. In such cases, attenuate the echo, for example by moving the sphere to a larger range than indicated in Figure 2.5.

### 2.1.2.5 Measurement environment

Measure $t_w$ and $s_w$ between the transducer and calibration sphere. These measurements are used to calculate the $c_w$ at the transducer and the sphere, e.g., using the equations in Leroy (1969), Del Grosso (1974), Mackenzie (1981), Fofonoff and Millard (1983), Leroy et al. (2008), or Figure 2.6, and the mean $c_w$ for the propagation path between the transducer and the sphere. The $c_w$ at the transducer should be used to calculate the equivalent two-way beam angle, $\psi$ (sr) (see Section 2.1.2.7.2). The $c_w$ at the sphere should be used to calculate the sphere $\sigma_{bs}$.

The mean sound speed for the propagation path should be calculated by weighting the sound speed values measured in each range (e.g. depth) increment by the time the wave spends in that range increment (Demer, 2004). This harmonic mean sound speed, $\bar{c}_w$, is calculated by weighting the $i$-th sound speed, $c_{w_i}$ (m s$^{-1}$), by the incremental time, $\Delta t_i$ (s), spent by the sound wave in the $i$-th range increment, $\Delta r_i = r_i - r_{i-1}$ (m) (Weinberg, 1971; De Moustier, 2001):

$$\bar{c}_w = \left( r_{\text{max}} - r_{\text{min}} \right) \left[ \sum_{i=1}^{N} \frac{1}{\Delta r_i} \log_{e} \left( \frac{c_{w_i} + 1}{c_{w_i}} \right) \right]^{-1},$$

where $g_i$ (s$^{-1}$) is the gradient $\partial c_w / \partial r$ in $\Delta r_i$, and $r_{\text{max}}$ (m) and $r_{\text{min}}$ (m) are the maximum and minimum ranges (e.g. depths), respectively. If $\partial c_w / \partial r_i$ is approximately zero, then the time spent in the $i$-th increment is calculated using a constant sound speed (De Moustier, 2001). If a constant sound speed is assumed for all of the increments, then Equation (2.3a) can be approximated by (Seabird, 2013):

$$\bar{c}_w \approx \frac{\sum_{i=1}^{N} \Delta r_i}{\sum_{i=1}^{N} r_i / c_{w_i}}.$$
Figure 2.6. Variation of sound speed with water temperature, $t_w$ (°C), and salinity, $s_w$ (psu, as indicated on the right axis). Calculated from the nine-term equation in (Mackenzie, 1981).

The measurements of $t_w$ and $s_w$ are also used to calculate the mean absorption coefficients, $\alpha_a$ (dB re 1 m$^{-1}$), for the propagation path, e.g. using equations in (Francois and Garrison, 1982a), Ainslie and McColm (1998), Doonan et al. (2003), or Figure 2.7 and 2.8.
2.1.2.6 Sphere positioning

Considering the geometry of the outriggers and transducer, calculate the line lengths required to place the sphere directly beneath the transducer, at \( r \geq 3r_{nf} \) (Equation 2.1b). Then, adjust the line lengths to position the sphere on the beam axis. This procedure is greatly facilitated by the angular coordinates of the sphere, \((\alpha, \beta)\), provided by a split-beam echosounder.

To position the sphere on the beam axis of a single-beam echosounder, the sphere position must be adjusted until the maximum intensity is measured. When the sphere is located on the beam axis, movement in any direction will result in lower echo intensity.

If a split-beam transducer is located in close proximity to a single-beam transducer, the split-beam sphere-position estimates may be translated, using knowledge of the relative transducer locations, to position the sphere in the centre of the single-beam (Demer et al., 1999; Simrad, 2001).

Measure \( \sigma_{bs} \) for the sphere on the beam axis. To expedite subsequent calibrations, mark or record the outrigger line lengths used to place the sphere in the centre of the beam for each transducer. Then, measure \( \sigma_{bs}(\alpha, \beta)^2 \) as the sphere is methodically move throughout the beam, to at least \( \pm \theta_{-3 \text{ dB}/2} \).

The sphere is efficiently moved throughout the beam by moving one line at a time. For example, place the sphere in the centre of the beam, then retrieve the line that is abeam of the transducer until the sphere reaches one edge of the beam. Release twice that amount of line until the sphere has reached the opposite edge of the beam. Then, shift the sphere forward or aft by shortening or lengthening the forward line and lengthening or shortening the aft line, respectively. Repeat the process with the line abeam,
moving the sphere to the opposite edge of the beam. Repeat the process until sphere- 
TS measurements are made throughout the beam.

Positioning the sphere may be difficult if the tethers are too long, particularly if the 
current is strong. In this case, add a ballast weight beneath the sphere (see Figure 2.3). 
Sphere positioning will be difficult if the outrigger geometry is incorrect. In this case, 
remeasure the geometry and adjust the positions of the outrigger(s) appropriately.

If part of the transducer is not transmitting or receiving, the sphere may not be detected 
even when it is physically located in portions of the beam. In this case, the sphere echos 
measured in the other parts of the beam will be weaker than expected, and the 
fit beam model will be significantly distorted. These symptoms may be caused by 
faulty electrical wiring between the transceiver and transducer or a faulty transducer 
or transceiver.

Current, vessel movement, animal echoes, or low tide can sometimes make it difficult 
to complete a calibration. In these situations, consider doing the calibration later, at a 
different location, or both. Persevering in difficult conditions often yields a poor quality 
calibration.

2.1.2.7 Sphere retrieval

After the calibration, update the echosounder and analysis software with the new cal-
ibration values (see Section 2.1.2.8). The calibration sphere may then be retrieved from 
under the vessel using the following steps:

1) Release line on the side with two outriggers until the outrigger line on the other 
side of the vessel hangs perpendicular to the water surface.

2) While retrieving the sphere and weight on the side with one outrigger, con-
tinue to release line on the side with two outriggers. Avoid touching the hull 
with the lines or calibration sphere as they can easily become entangled.

3) Untie the lines, temporarily stow the sphere and weight, and retrieve the other 
downrigger lines. If there is a current or if the lines are buoyant, a small weight 
may be attached to each of the lines before releasing them to keep them from 
catching on the hull while they are being retrieved.

4) Rinse the calibration sphere and weight in freshwater, allow them to dry, and 
store the sphere in its protective case.

2.1.2.8 Calibration analysis

2.1.2.8.1 Beam directivity

A split-beam echosounder estimates \( \alpha \) and \( \beta \), the alongships and athwartships target-
bearing angles, \( \theta \) (geometric°), from phase angles of electrical signals received by trans-
ducer sub-arrays, \( \varphi_e \) (electrical°):

\[
\varphi_e = k d_{eff} \sin(\theta) = A \sin(\theta),
\]

(2.4)

where \( A \) (electrical°/geometric°) is the transducer angle sensitivity, a function of wave-
length and thus \( c_w \). Bodholt (2002) showed that one estimate of angle sensitivity, \( A' \), 
measured at a certain sound speed, \( c_{w'} \), can be adjusted to the local sound speed, \( c_w \), 
using:

\[
A = A' \frac{c_{w'}}{c_w}.
\]

(2.5)
As the sphere is moved throughout the transducer beam, the split-beam angular coordinates, $\alpha$ and $\beta$ (°), can be used to measure $\sigma_{bb}(\alpha, \beta)^2$. For a real circular- or elliptical-beam transducer, $b(\alpha, \beta)^2$ can be modelled by:

$$b(\alpha, \beta)^2 = 10^{0.60206 \left( \frac{2\alpha}{\alpha_{-3\,\text{dB}}} \right)^2 + \frac{2\beta}{\beta_{-3\,\text{dB}}} - 0.18 \left( \frac{2\alpha}{\alpha_{-3\,\text{dB}}} \right)^2 \left( \frac{2\beta}{\beta_{-3\,\text{dB}}} \right)^2},$$  \tag{2.6}

where $\alpha_{-3\,\text{dB}}$ and $\beta_{-3\,\text{dB}}$ are the alongships and athwartships $-3$ dB beamwidths. Bodholt (2002) also showed that one estimate of beamwidth, $\alpha_{-3\,\text{dB}_0}$, measured at a certain sound speed, $c_{w'}$, can be adjusted to the local sound speed, $c_w$, using:

$$\alpha_{-3\,\text{dB}} = \alpha_{-3\,\text{dB}_0} \frac{c_w}{c_{w'}}, \quad \text{and} \quad \beta_{-3\,\text{dB}} = \beta_{-3\,\text{dB}_0} \frac{c_w}{c_{w'}} \tag{2.7a}$$

$$\beta_{-3\,\text{dB}} = \beta_{-3\,\text{dB}_0} \frac{c_w}{c_{w'}} \tag{2.7b}$$

The values for $\alpha_{-3\,\text{dB}}$ and $\beta_{-3\,\text{dB}}$ can be estimated from a regression of Equation (2.6) and the measures of $\sigma_{bb}(\alpha, \beta)^2$. It is important to exclude echoes from other sources (e.g. fish) and noise (e.g. crosstalk from other echosounders), as these can bias the model regression.

Proper system function is indicated by an undistorted beam model that has small deviations from the measurements (e.g. Figures 2.9 and 2.10). Conversely, if one quadrant of a circularly-symmetric, split-beam transducer is not functioning, then the beam model will be more oval than circular, and the system gain will be significantly different from the nominal value that is indicated by the manufacturer.

![Figure 2.9. Example of the beam model for a 7°-beamwidth, 38-kHz transducer, estimated using TS measurements of a 60.0-mm diameter copper sphere. Indications of a good quality calibration are uniform coverage of the sphere-TS measurements (+), and smooth, circular, and concentric contours. The colours indicate the observed (beam uncompensated) sphere TS (dB).](image)
Calibration of acoustic instruments

Figure 2.10. Example of TS measurements of a 60.0-mm diameter copper sphere (dots) and the fitted beam model for the 7°-beamwidth 38-kHz transducer (solid line). The orientation of each beam cross section is indicated in the legend. Note that the sphere TS measurements near the beam axis may be lower than the model of $b(\alpha, \beta)^2$ in Equation (2.6) (Pedersen, 2007).

Because both $\Lambda$ and $\theta_{-3\text{dB}}$ vary inversely with $c_w$, the directivity pattern model, Equation (2.6), is insensitive to changes in $c_w$ (Bodholt, 2002). This assumes that $\sin(\theta_{-3\text{dB}}) \approx \theta_{-3\text{dB}}$, which is valid for small angles. However, if $\alpha$ and $\beta$ are calculated using nominal values for $\Lambda_\alpha$ and $\Lambda_\beta$, the regression estimates of $\alpha_{-3\text{dB}}$ and $\beta_{-3\text{dB}}$ are not estimates of the true beamwidths. Nevertheless, if they are used in Equation (2.6), the resulting $b(\alpha, \beta)^2$ can be used to correctly normalize measurements of $\sigma_{\text{bs}}$ to the beam axis:

$$\sigma_{\text{bs}} = \frac{P_{\text{ext}} 16\pi^2 \times 10^{-\text{a}\theta/5}}{P_{\text{ext}} b(\alpha, \beta)^2 \lambda^2}.$$  

(2.8)

2.1.2.8.2 Equivalent beam angle

Because $>99\%$ of the transmitted energy is in the main lobe (Simmonds, 1984a, 1984b), the equivalent two-way beam angle, $\psi$ (sr), can be estimated from accurate estimates of the transducer beamwidth (Urick, 1983),

$$\psi \approx (\alpha_{-3\text{dB}} \cdot \beta_{-3\text{dB}})/5800.$$  

(2.9)

Because the regression values of $\alpha_{-3\text{dB}}$ and $\beta_{-3\text{dB}}$ (see Section 2.1.2.7.1) are not estimates of the true beamwidths, they should not be used in Equation (2.9) to estimate $\psi$. Instead, values for $\psi$, adjusted for the local sound speed, $c_w$, may be estimated from the transducer specific value, $\psi'$, measured by the echosounder manufacturer at sound speed, $c_w'$ (Bodholt, 2002):

$$\psi = \frac{\psi' c_{w}^2}{c_{w'}^2}.$$  

(2.10)
The transducer mounting may also affect $\psi$ (Simmonds, 1984b; Foote, 1987), but this affect is commonly and conveniently assumed to be negligible. Accurate estimates of $\psi$, $\Lambda_w$, $\beta$, $\alpha_{-3\text{dB}}$, and $\beta_{-3\text{dB}}$, all functions of $c_w$, require sphere-position estimates, $(r, \alpha, \beta)$, that are independent of those reported by a split-beam echosounder (Simmonds, 1984b; Reynisson, 1998). Such measurements require specialized equipment, a controlled environment, or both, and are generally not part of a sphere calibration.

Summarizing, it is recommended that a value for $\psi$ is estimated from the transducer specific measurement made by the manufacturer in controlled conditions (e.g. a test tank) at a known sound speed, $c_w^*$, adjusted for the local $c_w$ (Bodholt, 2002). Furthermore, it is recommended that nominal values are used for $\Lambda_w$ and $\beta$, and then $\alpha_{-3\text{dB}}$ and $\beta_{-3\text{dB}}$ are estimated from the sphere calibration results. If this procedure is followed, the beamwidth values may not be the true values, but the directivity-pattern compensation will be correct (Bodholt, 2002).

2.1.2.8.3 On-axis gain

The calibrated $g_0$ is calculated from the arithmetic mean of sphere-$\sigma_{bs}$ measurements made on the beam-axis, $\bar{\sigma}_{bs}$:

$$g_0 = \bar{g}_0 \left( \frac{\sigma_{bs}}{\bar{\sigma}_{bs}} \right)^{0.5}$$

(2.11)

where $\bar{g}_0$ is the uncalibrated $g_0$, and $\sigma_{bs}$ is that calculated for the sphere (see Section 2.1.1.1). Because it is often difficult to make sphere-$\sigma_{bs}$ measurements exactly on the beam axis, $\bar{\sigma}_{bs}$ is generally estimated as the arithmetic mean of the sphere-$\sigma_{bs}$ measurements made near the beam-axis, where $b(\alpha, \beta) \approx 1$, e.g. $\leq 0.5^\circ$. It is important to remove noise (e.g. echoes from swivels, biological scatterers, and the weight), as they can bias $\bar{\sigma}_{bs}$. Once $g_0$ has been calculated and entered into the echosounder, check that $\bar{\sigma}_{bs} \approx \sigma_{bs}$, within the echosounder’s measurement accuracy.

A large change in the calibrated gain, e.g. $\geq 12\%$, could be caused by (i) high acoustic absorption on the transducer face, perhaps due to bio-fouling or bubbles (follow the manufacturer’s recommendation for cleaning the transducers); (ii) high acoustic absorption between the transducer and the sphere, perhaps due to air bubbles or suspended organic matter; or (iii) a damaged transducer, transceiver, cable, or connector (in some cases, the directivity pattern may still appear to be correct).

2.1.3 Alternative platforms

Depending on the transducer platform, the three-line method may be modified or replaced by another method. For example, to calibrate a system with a horizontally-projecting transducer, the sphere may be moved within the beam using two reels and outriggers. To calibrate a pole-mounted system, it may be necessary to extend the outrigger nearest to the transducer pole. If this is not possible, the calibration sphere may be positioned below the transducers using a single monofilament line. In this case, the sphere may be moved within the acoustic beams intentionally by moving the line or pseudo-randomly as a result of vessel motion.

For some platforms, it may be more appropriate to hold the sphere relatively stationary (e.g. suspended from one or more lines) while the transducer is pitched and rolled. For platforms with vertically-projecting transducers, it is sometimes advantageous to suspend the sphere in the beam throughout the measurements, e.g. buoyed above a lander with a float or by gravity beneath a towed body. The latter method provides concurrent
measurements of $\sigma_{bs}$ from the sphere and targets and monitors changes in $g_0$ vs. $d_w$ or $p_w$ (Dalen et al., 2003) (e.g. Figure 2.11). However, forward scatter from the sphere can affect the measurements of the targets (MacLennan, 2011), and vice versa. These effects can be reduced by increasing the range between the sphere and biological targets.

Figure 2.11. Near-axis [$b(\alpha, \beta)^2 \leq 0.5$ dB] measurements of target strength, $T_S$ (dB re 1 m$^2$), made with a constant $g_0$ setting in the echosounder (Simrad EK60), from a 38.1-mm WC sphere suspended below a 120-kHz transducer (Simrad 120-7DD) during one descent to over 800-m depth. Overlaid is a 5th-order polynomial fit to the data (black line) with upper and lower 95% confidence intervals (grey lines).

Transducers mounted in deep-towed bodies should be designed specifically for deep deployments. Air-backing or rubber faces on some transducers will exhibit hysteresis in $g_0$ vs. $d_w$ (e.g. see Figure 8.1 in Ona, 1999). Such transducers should be calibrated and operated within a narrow range of shallow depths.

For echo-integration-based studies with a deeply deployed transducer, $\psi$ should also be quantified vs. $d_w$. Foote (1987) and Bodholt (2002) described the dependence of $\psi$ on $c_w$, which generally changes with $d_w$, but further work is needed to determine how $\psi$ changes vs. $p_w$.

Optimally, calibrations should be conducted with the echosounder system installed in the survey platform. If this is not possible, they may be calibrated in a controlled environment, e.g. in a test tank, and separately from the platform (Scalabrin et al., 2009).
3 Calibration uncertainty

3.1 Introduction

There are two types of uncertainty in measurement of any quantity: (i) precision (random error) and (ii) accuracy (systematic error or bias). The former indicates the relative variability, or repeatability, of the measurements; the latter indicates the closeness of the measurements to the true value (Figure 3.1). This chapter explores uncertainty in acoustic system calibrations.

![Figure 3.1. Illustration of accuracy and precision. The measurements in this example have a normal or Gaussian distribution with mean = 2.5 and standard deviation = 0.44. The deviation of the mean from the true value represents the measurement accuracy (bias or systematic error). The standard deviation represents the measurement precision (variability or random error).](image)

3.2 Random error

Random noise resulting from system electronics (electric noise) and ambient conditions [acoustic noise; Wenz (1962)] may cause random error in echosounder measurements. This noise may be measured with the echosounder operating in passive mode without transmitting pulses. Calibration spheres are selected to have $\sigma_{bs}$ values such that the signal-to-noise-ratio at the calibration site is larger than 100 (20 dB). In this case, ambient noise is generally a negligible source of calibration error.

Random error in measurements of beam-compensated sphere $\sigma_{bs}$ is often estimated as the root-mean-square (rms) deviation of the measured sphere $\sigma_{bs}$ from the directivity-pattern model. With good practice and accounting for the environment, modern scientific echosounders may have calibrated $g_0$ values that are precise to about ±0.1 dB over multiple years (Simmonds, 1990; Knudsen, 2009). Although this estimate may increase with increasing frequency, the random error due to calibration is likely a minor contributor to the total error in an acoustic survey (Demer, 2004; O’Driscoll, 2004; Loland et al., 2007; Woillez et al., 2009).
3.3 Systematic error

The systematic error or accuracy in a system calibration depends foremost on the accuracy of the backscattering cross section, $\sigma_{bs}$ (m$^2$), estimated for the calibration sphere. It also depends on the accuracy of many other parameters in Equation (1.12b). Furthermore, each of these parameters and, therefore, the system response may change after the system is calibrated, e.g. due to changes in the operating environment (Demer, 2004; Demer and Renfree, 2008). In the following subsections, we explore the relative magnitudes of calibration uncertainties potentially contributed by these parameters.

3.3.1 Sphere target strength

First explored is the accuracy of sphere $\sigma_{bs}$, based on a theoretical model of backscattering from solid elastic spheres (Faran, 1951; MacLennan, 1981). A Matlab (The Mathworks, USA) implementation of the model is available. The model is exact, so the accuracy of the computed $\sigma_{bs}$ depends on the accuracy of the estimated physical and geometric model parameters relative to the actual sphere.

3.3.1.1 Material properties

The material properties of an elastic sphere include the density of the sphere, $\rho_s$ (kg m$^{-3}$), the compressional (or longitudinal) wave sound speed, $c_c$ (m s$^{-1}$), and the shear (or transverse) wave sound speed, $c_s$ (m s$^{-1}$). Nominal values for tungsten carbide with 6% cobalt binder (WC) and copper (Cu) are listed in Table 3.1.

<table>
<thead>
<tr>
<th>Property</th>
<th>Symbol</th>
<th>WC</th>
<th>Cu</th>
</tr>
</thead>
<tbody>
<tr>
<td>Sphere material density</td>
<td>$\rho_s$ (kg cm$^{-3}$)</td>
<td>14.90×10$^{-3}$</td>
<td>8.95×10$^{-3}$</td>
</tr>
<tr>
<td>Sphere compressional wave sound speed</td>
<td>$c_c$ (m s$^{-1}$)</td>
<td>6 853.0</td>
<td>4 760.0</td>
</tr>
<tr>
<td>Sphere shear wave sound speed</td>
<td>$c_s$ (m s$^{-1}$)</td>
<td>4 171.0</td>
<td>2 288.5</td>
</tr>
<tr>
<td>Sphere depth</td>
<td>$d_w$ (m)</td>
<td>30</td>
<td>30</td>
</tr>
<tr>
<td>Water temperature</td>
<td>$t_w$ (°C)</td>
<td>10</td>
<td>10</td>
</tr>
<tr>
<td>Water salinity</td>
<td>$s_w$ (psu)</td>
<td>35</td>
<td>35</td>
</tr>
<tr>
<td>Percent bandwidth</td>
<td>$%f$ (%)</td>
<td>1.0</td>
<td>1.0</td>
</tr>
</tbody>
</table>

The variation of theoretical sphere target strength, $TS = 10\log_{10}(\sigma_{bs})$ (dB re 1 m$^2$), was simulated for a 38.1-mm-diameter sphere made from tungsten carbide with 6% cobalt (Figure 3.2). In computations for six frequencies, $f$ (Hz), that are commonly used in fishery acoustics surveys, the $\rho_s$, $c_c$, and $c_s$ were each varied within ±4% of their nominal value (Table 3.2) while the other parameters were held constant at their nominal values (Table 3.1).
Figure 3.2. Theoretical target strength (TS, dB re 1 m²) vs. frequency, \( f \) (Hz), for a 38.1-mm-diameter sphere made of tungsten carbide with 6% cobalt, calculated with ±4% variations in (a) density, \( \rho_s \) (kg m\(^{-3}\)), (b) compressional-wave sound speed, \( c_c \) (m s\(^{-1}\)), and (c) shear-wave sound speed, \( c_s \) (m s\(^{-1}\)). Nominal values (Table 3.1) were used for the other parameters. Because there is a null near \( f = 200 \) kHz, the minimum TS, calculated with \( c_c = 1.04 \), is >13 dB lower than the nominal TS.

Variation in \( \rho_s \) affects the mean \( \sigma_{bs} \), but it has relatively little effect on the normalized frequency response, \( \sigma_{bs}(f) / \sigma_{bs} \), e.g., the positions of the peaks and nulls (Figure 3.2a). The \( c_c \) and \( c_s \) values affect both \( \overline{\sigma_{bs}} \) and \( \sigma_{bs}(f) / \sigma_{bs} \) values (Figure 3.2b-c). The \( \sigma_{bs} \) values are most sensitive to changes in \( c_s \), and this sensitivity increases with \( f \).
Table 3.2. Theoretical target strength (TS; dB re 1 m²) vs. material properties for a 38.1-mm-diameter sphere made of tungsten carbide with 6% cobalt, calculated with ±4% variations in density, \( \rho_s \) (kg m\(^{-3} \)), (b) compressional-wave sound speed, \( c_c \) (m s\(^{-1} \)), and (c) shear-wave sound speed, \( c_s \) (m s\(^{-1} \)). Nominal values (Table 3.1) were used for the other parameters and to calculate the nominal TS. Values for TS range = maximum TS – minimum TS ≥0.1 dB re 1 m² (≥2.3% variation) are bold.

<table>
<thead>
<tr>
<th>Parameter</th>
<th>Frequency (Hz)/1 000</th>
</tr>
</thead>
<tbody>
<tr>
<td>Density, ( \rho_s )</td>
<td></td>
</tr>
<tr>
<td>Minimum</td>
<td>-42.75</td>
</tr>
<tr>
<td>Nominal</td>
<td>-42.70</td>
</tr>
<tr>
<td>Maximum</td>
<td>-42.66</td>
</tr>
<tr>
<td>Range</td>
<td>0.09</td>
</tr>
<tr>
<td>Compressional speed, ( c_c )</td>
<td></td>
</tr>
<tr>
<td>Minimum</td>
<td>-42.70</td>
</tr>
<tr>
<td>Nominal</td>
<td>-42.70</td>
</tr>
<tr>
<td>Maximum</td>
<td>-42.70</td>
</tr>
<tr>
<td>Range</td>
<td>0.00</td>
</tr>
<tr>
<td>Shear speed, ( c_s )</td>
<td></td>
</tr>
<tr>
<td>Minimum</td>
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<tr>
<td>Nominal</td>
<td>-42.70</td>
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<tr>
<td>Maximum</td>
<td>-42.70</td>
</tr>
<tr>
<td>Range</td>
<td>0.00</td>
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</tbody>
</table>

3.3.1.2 Sphericity

The accuracy in machining or sintering the sphere is normally better than 0.1 mm. The acoustic wavelength is >4 mm for frequencies <350 kHz. Therefore, the sphericity of the calibration target likely has negligible effect on the sphere TS.

3.3.1.3 Temperature, salinity, and pressure

Sphere \( \sigma_{ts} \) varies with temperature because \( c_c, c_s, \rho_s \), and water sound speed, \( c_w \) (m s\(^{-1} \)), are functions of temperature (more discussion in Section 3.2.10.1). However, the temperature dependences of \( c_c \) and \( c_s \) are not well characterized. Shown are the TS values of a 38.1-mm-diameter sphere made from tungsten carbide with 6% cobalt vs. \( t_w = 5, 10, \) and 20°C; water salinity, \( s_w = 0, 30, 33, \) and 35 psu; and water pressure = 0, 10, 30, and 100 dbar (Table 3.3).

Table 3.3. Theoretical target strength (TS; dB re 1 m²) vs. water temperature, \( t_w = 5, 10, \) and 20°C; water salinity, \( s_w = 0, 30, 33, \) and 35 psu; and water pressure, \( p_w = 0, 10, 30, \) and 100 dbar for a 38.1-mm-diameter sphere made from tungsten carbide with 6% cobalt. Nominal values (Table 3.1) were used for the other parameters. Range values ≥0.1 dB re 1 m² (≥2.3% variation) are bold.

<table>
<thead>
<tr>
<th>Parameter</th>
<th>Frequency (Hz)/1 000</th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
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</tr>
<tr>
<td>Temperature, ( t_w )</td>
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</tr>
<tr>
<td>5</td>
<td>-42.87</td>
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<td>10</td>
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<tr>
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<td>0</td>
<td>-43.08</td>
</tr>
<tr>
<td>30</td>
<td>-42.75</td>
</tr>
</tbody>
</table>
### 3.3.1.4 Measurement bandwidth

The bandwidth of a pulsed sinusoidal signal, \( b_f \) (Hz), is approximately equal to the inverse of the pulse duration, \( \tau \) (s). The sphere \( \sigma_{bs} \) should be calculated and averaged over \( b_f \) before being expressing as target strength (\( TS; \text{dB re 1 m}^2 \)) (Table 3.4).

<table>
<thead>
<tr>
<th>Bandwidth (%( f ))</th>
<th>Frequency (Hz)/1000</th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
<td>18</td>
</tr>
<tr>
<td>0</td>
<td>-42.70</td>
</tr>
<tr>
<td>0.5</td>
<td>-42.70</td>
</tr>
<tr>
<td>1.0</td>
<td>-42.70</td>
</tr>
<tr>
<td>1.5</td>
<td>-42.70</td>
</tr>
<tr>
<td>2.0</td>
<td>-42.70</td>
</tr>
<tr>
<td>Range</td>
<td>0.00</td>
</tr>
</tbody>
</table>

Table 3.4. Theoretical target strength (\( TS; \text{dB re 1 m}^2 \)) vs. bandwidth, percent of frequency, \( f \) (Hz), for a 38.1-mm-diameter sphere made of tungsten carbide with 6% cobalt. The model parameters were nominal values (Table 3.1). The receiver bandwidth is assumed to be much larger than \%\( f \). Values for \( TS \) Range \( \geq 0.1 \text{ dB re 1 m}^2 \) (\( \geq 2.3\% \) variation) are bold.

### 3.3.1.5 Sphere suspension

The contribution of the sphere suspension line to the uncertainty of the \( TS \) measurement depends on its thickness, the sphere attachment method, and the echosounder frequency. The monofilament line should have a diameter that is much smaller than the wavelength; it should be as thin as possible to support the sphere and perhaps a stabilizing weight. The knots should be at least \( c_w \tau \) metres from the sphere. Typically, the uncertainty introduced by the tether is negligible at 18 and 38 kHz, but increases with frequency and is significant (~0.5 dB or more) at 200 kHz and higher.

### 3.3.2 Sound speed and absorption

The received power is compensated, as a function of propagation range, for attenuation due to spherical spreading and absorption. The range estimate is a function of sound speed, which is a function of temperature, salinity, and pressure. The equations used to estimate sound speed values are mostly heuristic and empirical.

There are a number of algorithms for estimating sound speed in seawater (Wilson, 1960; Del Grosso, 1974; Chen and Millero, 1977; Mackenzie, 1981; Speisberger and Metzger, 1991; Millero and Li, 1994; Leroy and Parthiot, 1998), but the algorithm proposed by Chen and Millero (1977) is used most (Fofonoff and Millard, 1983). It is endorsed by the UNESCO/SCOR/ICES/IAPSO Joint Panel on Oceanographic Tables and Standards and SCOR Working Group 51. However, Pike and Beiboer (1993) compared...
Calibration of acoustic instruments

algorithms and recommended the Chen and Millero (1977) algorithm for water depth <1000 m, and the Del Grosso (1974) algorithm for water depth >1000 m. The two algorithms differ mainly due to pressure (Table 3.5).

Table 3.5. Sound speed in seawater calculated using algorithms by Chen and Millero (1977) and Del Grosso (1974) at different pressures.

<table>
<thead>
<tr>
<th>Pressure (dbar)</th>
<th>Chen and Millero (ms⁻¹)</th>
<th>Del Grosso (ms⁻¹)</th>
<th>Difference (ms⁻¹)</th>
</tr>
</thead>
<tbody>
<tr>
<td>0</td>
<td>1.489.79</td>
<td>1.489.82</td>
<td>-0.03</td>
</tr>
<tr>
<td>10</td>
<td>1.491.38</td>
<td>1.489.99</td>
<td>1.40</td>
</tr>
<tr>
<td>50</td>
<td>1.497.78</td>
<td>1.490.65</td>
<td>7.14</td>
</tr>
<tr>
<td>100</td>
<td>1.505.82</td>
<td>1.491.47</td>
<td>14.35</td>
</tr>
<tr>
<td>500</td>
<td>1.571.36</td>
<td>1.498.07</td>
<td>73.29</td>
</tr>
<tr>
<td>1000</td>
<td>1.654.84</td>
<td>1.506.34</td>
<td>148.50</td>
</tr>
<tr>
<td>1500</td>
<td>1.737.52</td>
<td>1.514.64</td>
<td>222.88</td>
</tr>
</tbody>
</table>

Sound absorption in seawater results from shear and bulk viscosities, and chemicals (Thorpe, 1967; Fisher and Simmons, 1977; Francois and Garrison, 1982a, 1982b; Ainslie and McColm, 1998; Doonan et al., 2003). The sound absorption coefficient for seawater \( t = 4^\circ C, s = 35 \text{ psu}, p = 300 \text{ dbar}, pH = 8, \text{ Lat} = 48^\circ N \) is calculated vs. \( f \) using multiple algorithms (Table 3.6). The algorithm by Francois and Garrison (1982a) is probably the most used, but other formulae were validly obtained by fitting different sets of measured data.

The algorithm by Ainslie and McColm (1998) is a simplified version of that by Francois and Garrison (1982a), so it gives essentially the same results. Doonan et al. (2003) reprocessed the data in Francois and Garrison (1982a) and, ignoring the boric acid component, proposed another algorithm that gives essentially the same results for \( f \) between 10 and 120 kHz, and \( t_w < 20^\circ C \). However, values computed by other algorithms (Thorpe, 1967; Fisher and Simmons, 1977) differ more, e.g. >1.5 dB km⁻¹ at 38 kHz and >4.5 dB km⁻¹ at 200 kHz.

Table 3.6. Seawater absorption coefficients, \( \alpha_a \) (dB km⁻¹), computed with algorithms from various authors, each parameterized with: \( t_w = 4^\circ C, s_w = 35 \text{ psu}, p_w = 300 \text{ dbar}, pH = 8, \text{ Lat} = 48^\circ N \).

<table>
<thead>
<tr>
<th>Reference</th>
<th>Frequency (kHz)</th>
</tr>
</thead>
<tbody>
<tr>
<td>18</td>
<td>38</td>
</tr>
<tr>
<td>Francois and Garrison (1982a)</td>
<td>3.15</td>
</tr>
<tr>
<td>Ainslie and McColm (1998)</td>
<td>3.19</td>
</tr>
<tr>
<td>Doonan et al. (2003)</td>
<td>2.67</td>
</tr>
<tr>
<td>Fisher and Simmons (1977)</td>
<td>2.35</td>
</tr>
<tr>
<td>Thorpe (1967)</td>
<td>3.43</td>
</tr>
</tbody>
</table>

3.3.3 Directivity pattern

The directivity pattern, \( b(\alpha, \beta) \) (dimensionless), and equivalent beam angle, \( \psi \) (sr), vary as the ratio of the transducer size to the acoustic wavelength, \( \lambda \) (m). The latter varies with sound speed, \( c_w \) (m s⁻¹). Therefore, as sound speed changes from one value, \( c_w \), to another, \( c_w' \), the beamwidth and equivalent beam angle change, the transmit power is more or less focused, and the on-axis gain, \( g_0 \) (dB re 1), is changed by the
factor \(\left(\frac{c_w}{c_{w0}}\right)^2\) (Bodholt, 2002). For example, if \(c_w\) changes from 1500 to 1450 m s\(^{-1}\), \(g_0\) will change by 7%.

### 3.3.4 Equivalent beam angle

Normally, \(\psi\) is expressed in decibels, \(\psi\) (dB re 1 sr). Assuming backward radiation is negligible and the signal-to-noise ratio, \(r_{sn}\) (dimensionless), is infinite, \(\psi\) is calculated by integrating \(b(\alpha, \beta)^2\) over a hemisphere (half space). Realistically, \(r_{sn}\) is less than ideal, and the integration should, therefore, be performed over the space corresponding to the actual \(r_{sn}\). Values of \(\psi\) were computed for a circular piston transducer with nominal beamwidths ranging from 6.6° to 7.4°, and integration spaces corresponding to various \(r_{sn}\) values (Table 3.7).

#### Table 3.7. Equivalent beam angle, \(\psi\) (dB re 1 sr), vs. beamwidth \(\theta_{-3\text{ dB}}\) (°) and the integration space, which depends on the signal-to-noise ratio, \(r_{sn}\) (dimensionless).

<table>
<thead>
<tr>
<th>(\theta_{-3\text{ dB}}) (°)</th>
<th>Integration space ((r_{sn}))</th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
<td>3 dB</td>
</tr>
<tr>
<td>7.0</td>
<td>-21.92</td>
</tr>
</tbody>
</table>

### 3.3.5 Linearity and dynamic range

In fishery acoustics applications, the term linearity confusingly refers to (i) the summation of echoes from individual targets in an observation volume (Foote, 1983a), (ii) the echosounder-receiver response over its dynamic range, or (iii) the propagation of high-intensity sound pulses (Tichy et al., 2003; Pedersen, 2007; Korneliussen et al., 2008). The latter two are relevant to acoustic system calibrations.

In the linear region of the echosounder response, the output values are proportional to the input values. Uncertainty in this proportionality is commonly expressed as the deviation from a linear input–output relationship, expressed as a percentage of full scale. Because sphere calibrations are typically conducted with one transmit power and one sphere at one range, they are only valid for the linear region that was calibrated.

Dynamic range is the ratio of the maximum to minimum possible values of a variable quantity. Optimally, the linear range spans the dynamic range. When calibrating with targets at short ranges, particularly when conducting a self-reciprocity calibration, it is important to assure that the echo signals are not outside the dynamic range, i.e. the echosounder receiver is not saturated, and the measurements are in the linear region.

When sound propagation is non-linear, power is shifted from the signal frequency to its harmonics. This can occur when the power density is too high. When this occurs, the echosounder response is nonlinear. To avoid non-linear propagation, use sufficiently low transmit-power densities (Table 4.5) [Table 1 in Korneliussen et al. (2008)].

### 3.3.6 System stability

System gain may change between the time an echosounder is calibrated and used, due to changes in the environmental and system parameters (Van Buren et al., 1999). For example, Demer and Renfree (2008) demonstrated that temperature can change the impedance and thus performance of some transducers. For applications involving a
towed or cast acoustic system, the on-axis system gain and directivity pattern may change significantly with changes in pressure (Beamiss et al., 2002). The performance of system electronics and transducers may change over time, gradually or abruptly (Blue, 1984). Biofouling over the course of a survey may cause signal attenuation and degrade the calibration accuracy. Therefore, it is important to characterize such changes and apply any necessary corrections. A sudden change in system performance indicates a problem with the system. Frequent calibrations will monitor for gradual changes.

3.4 Total error

3.4.1 Backscattering cross section

To assess the systematic error or bias in on-axis gain, \( g_0 \) (dimensionless), due to bias in the aforementioned parameters, rearrange Equation (1.12c), and substitute \( r = c_w t_r / 2 \), where \( t_r \) (s) is the time corresponding to scattering at range, \( r \) (m):

\[
g_0 = \frac{\pi f_w t_r^2 \omega_w^2 r_f^{20}}{\sigma_0 \cdot \delta t_0^{0.5}}.
\]

Ignoring parameter covariance, the relative bias in \( g_0 \) can be estimated by:

\[
\frac{\Delta g_0}{g_0} = \frac{\Delta f}{f} + \frac{\Delta c_w}{c_w} + \frac{2 \Delta t_r}{t_r} - \frac{\Delta \sigma_0}{\sigma_0} - \frac{\Delta \delta t_0}{\delta t_0} + \log_e(10) \left( \frac{\Delta c_w t_r \sigma_0}{2 \delta t_0} + \frac{c_w \Delta c_w t_r}{2 \sigma_0} + \frac{\Delta c_w t_r \delta t_0}{2 \sigma_0} \right).
\]

Grouping similar terms, we obtain:

\[
\left| \frac{\Delta g_0}{g_0} \right| \leq \frac{|\Delta \sigma_0|}{\sigma_0} + \frac{\log_e(10)}{20} c_w t_r |\Delta c_w| + \left( \frac{\log_e(10)}{20} \sigma_0 t_r + \frac{1}{c_w} \right) |\Delta c_w| \\
+ \frac{|\Delta f|}{f} + \frac{|\Delta \sigma_0|}{\sigma_0} + \frac{|\Delta t_r|}{t_r}.
\]

To estimate the relative bias in \( g_0 \) after a system calibration with a sphere, solve Equation (3.3) using a relevant scenario, perhaps assuming that the:

1) sphere is manufactured to specifications;
2) sphere material is homogeneous (or any deviations are negligible);
3) sphere is perfectly spherical (or any deviations are negligible);
4) sphere materials change negligibly vs. pressure (or depth);
5) effect of the sphere tether is negligible;
6) sound propagation is linear;
7) echosounder response is linear; and
8) environmental conditions vary as those encountered during ocean surveys.

For example, consider the following hypothetical scenario:

The difference in the absorption coefficient, \( \Delta \alpha_s \), is the difference between the values from (Francois and Garrison, 1982a) and those from Ainslie and McColm (1998), with the parameters given in Table 3.6. The \( \sigma_{bs} \) is for a 38.1-mm-diameter sphere made from tungsten carbide with 6% cobalt; it is a function of \( t_w, s_w, p_w, \rho_w, c_w, c_s, \) and \( b_f \):

\[
\Delta \sigma_{bs} = \max \left[ \sigma_{bs}(t_w, s_w, p_w, \rho_w, c_w, c_s, b_f) - \sigma_{bs}(t_w, s_w, p_w, \rho_s, c_c, c_s, b_f) \right],
\]

where each parameter and its prime counterpart indicate the values within that parameter’s expected range which yield, in combination with those values for the other pa-
rameters, the maximum difference in $\sigma_{bs}$ values, $\Delta \sigma_{bs}$. Values for $\rho_s$, $c_c$, $c_v$ were varied $\pm 2\%$ about their nominal values (Table 3.1); $t_w = 10 \pm 5^\circ C$; $b_f = 1$ kHz; $\Delta t_r = 1 \mu s$; $\Delta f / f = 0.01$; and $\Delta p_{cr} / p_{cr} = \Delta p_{st} / p_{st} = 0.01$.

The results of this simulation, conditioned by the aforementioned assumptions and scenario, indicates that the uncertainty in $g_0$ is mostly due to relative bias in $\sigma_{sa}$ and $\sigma_{cs}$, then $c_w$. Their relative contributions depend on $f$ (Table 3.8). Because co-occurrence of extreme relative bias in each parameter is improbable, the distribution of total relative bias in $g_0$ should be calculated by estimating the distributions of each parameter and solving Equation (3.3) by Monte Carlo simulation (Demer, 1994, 2004; Rose, 2003; O’Driscoll, 2004). This was not done here due to a lack of knowledge about the parameter distributions.

Table 3.8 Components of relative bias for gain, $g_0$ (dimensionless), vs. frequency, $f$ (Hz). The total relative bias in $g_0$ should be calculated by estimating the distributions of each parameter and solving Equation (3.3) by Monte Carlo simulation (Demer, 1994, 2004; Rose, 2003; O’Driscoll, 2004). Measurements of backscattering cross section, $\sigma_{bs}$ (m$^2$ m$^{-3}$), involve $g_0^2$, and, therefore, have twice the relative bias for $g_0$.

<table>
<thead>
<tr>
<th>Relative uncertainty</th>
<th>Frequency (Hz) / 1 000</th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
<td>18</td>
</tr>
<tr>
<td>$</td>
<td>\Delta \sigma_{bs}</td>
</tr>
<tr>
<td>$</td>
<td>\Delta \sigma_{sa}</td>
</tr>
<tr>
<td>$</td>
<td>\Delta \sigma_{cs}</td>
</tr>
<tr>
<td>$</td>
<td>\Delta c_w</td>
</tr>
<tr>
<td>$</td>
<td>\Delta p_{cr}</td>
</tr>
<tr>
<td>$</td>
<td>\Delta t_r</td>
</tr>
</tbody>
</table>

### 3.4.2 Volume backscattering coefficient

The calibration for measurements of volume backscattering coefficient, $s_v$ (m$^2$ m$^{-3}$), involves both $g_0^2$ and a correction factor, $s_{a \ corr}^2$. Comparing Equations (1.12b) and (1.16) and substituting $r = c_w t_r / 2$ and Equation (1.13):

$$s_v = \sigma_{bs} \frac{2}{s_{a \ corr}^2 c_w \psi r^2} = \sigma_{bs} \frac{8}{t_{eff} c_w \psi r^2} = \sigma_{bs} k_{sp} r,$$

(3.5)

where $k_{sp}$ is a substitute for the quotient $B / t_{eff} c_w \psi r^2$. Therefore, the relative bias in $s_v$, $\Delta s_v$, includes the relative bias in $g_0^2$, i.e. twice the values shown in Equation (3.3), plus the relative bias in $k_{sp}$, which can be estimated by:

$$\left| \frac{\Delta k_{sp}}{k_{sp}} \right| \leq \frac{|\Delta t_{eff}|}{t_{eff}} + \frac{|\Delta c_w|}{c_w} + \frac{|\Delta \psi|}{\psi} + \frac{2|\Delta t_r|}{t_r}.$$

(3.6)

This equation may also underestimate the total relative bias because it does not account for parameter covariance. A hypothetical scenario could include the following values for the parameters in Equation (3.5): $|\Delta t_{eff}| / t_{eff} = 0.05$, $|\Delta c_w| / c_w = 20$ m$^{-1}$, $|\Delta \psi| / \psi = 0.07$ (Table 3.7), and $\Delta t_r = 1 \mu s$. The distribution of total relative bias should be calculated by estimating the distributions of each parameter and solving Equations (3.3) and (3.6) by Monte Carlo simulation (Demer, 1994, 2004; Rose, 2003; O’Driscoll, 2004).
that case, the relative bias in calibration for measures of $s_v$ can be approximated as
$$2|\Delta g_0/g_0| + |\Delta k_{s_v}/k_{s_v}|.$$
4 Calibration protocols

This section presents calibration protocols for several commonly-used acoustic instruments including scientific and commercial echosounders (Simrad EK60, ES60, and ES70) and a multibeam sonar (Simrad ME70). Protocols are presented for systems with vessel-mounted transducers, either mounted on the hull or in a centreboard. With minor modification, these protocols can be adapted to systems with other transducer-mount configurations (see Section 1.4).

4.1 Simrad EK60, ES60, and ES70

4.1.1 Introduction

The EK60 scientific echosounder (Simrad, 2008) replaced the EK500 (Simrad, 1997) in 2003. Hereafter, EK60 refers to a complete single-frequency echosounder system including the General Purpose Transceiver (GPT), connectors and cable, transducer, computer, and the ER60 control and data-logging software (Simrad, 2008). Calibration.exe refers to the ER60 calibration utility that may be run from, or independently from, the ER60.

The ES60 commercial echosounder (Simrad, 2004) and its successor, the ES70 (Simrad, 2010), are composed of EK60-generation transceiver hardware, but have different firmware, software, and features. They too can be calibrated and used effectively in certain research situations. ES60 and ES70 refer to the respective echosounder systems and their software with the same names.

Most of the calibration protocols presented in this document apply generically to EK60, ES60, and ES70 systems. System-specific procedures are highlighted. Protocols are presented for both single-beam and split-beam systems.

These echosounders are most commonly calibrated using the sphere method (Chapter 2). The results of a sphere calibration are relevant to the entire echosounder system. Therefore, if the GPT, transducer, cable, or connector are changed or altered, the system should be recalibrated. The parameters to be quantified for a calibrated system are shown in Table 4.1.

<table>
<thead>
<tr>
<th>Parameter</th>
<th>Symbol</th>
<th>Units</th>
<th>Purpose</th>
<th>Description</th>
<th>Source</th>
</tr>
</thead>
<tbody>
<tr>
<td>On-axis gain</td>
<td>$G_0$</td>
<td>dB re 1</td>
<td>Estimating target strength</td>
<td>On-axis efficiency</td>
<td>Measured</td>
</tr>
<tr>
<td>$s_a$ correction</td>
<td>$S_{a,corr}$</td>
<td>dB re 1</td>
<td>Estimating volume backscatter coefficients ($s_v$) and integrated $s_v$</td>
<td>Difference in energy of the nominal and actual received pulses</td>
<td>Measured</td>
</tr>
<tr>
<td>Beamwidth</td>
<td>$\alpha_{-3,\text{dB}}$ and $\beta_{-3,\text{dB}}$</td>
<td>Degree</td>
<td>Estimating TS and school dimensions</td>
<td>Angle between half-power points</td>
<td>Nominal, adjusted for local $c_w$, or measured</td>
</tr>
</tbody>
</table>
The ER60 Calibration.exe utility is used to model the transducer directivity pattern and thereby provide estimates of on-axis gain, $G_0$ (dB re 1), $s_a$ correction, $S_a \text{corr}$ (dB re 1), 3-dB beamwidths, $\alpha_{-3 \, \text{dB}}$ and $\beta_{-3 \, \text{dB}}$ (°), and angle offsets, $\alpha_o$ and $\beta_o$ (°). In this chapter, the advantages and limitations of this application, and alternative approaches, are presented.

### 4.1.1.1 Data

The EK60, ES60, and ES70 echosounder systems output data in two formats that can be used for quantitative acoustic measurements: (i) raw data (.raw); and, converted to volume backscattering strength, $S_v$ (m$^2$ m$^{-3}$) and resampled, (ii) legacy (EK500) or ek5 data (.ek5 or Q-telegram).

For each echosounder transmission (ping), the raw data include received echo power, $P_{et}$ (W), phase angles in two orthogonal planes, $\alpha$, $\beta$ (°) (for split-beam systems), and the following GPT settings: frequency, $f$ (kHz), transmit power, $p_{et}$ (W), pulse duration, $\tau$ (s), $g_0$, $S_a \text{corr}$, and equivalent two-way beam angle, $\Psi$ (dB re 1 sr). These data and settings are used to calculate and display $S_v$ and target strength, $TS$ (dB re 1 m$^2$):

\[
S_v = P_{er} + 20 \log_{10}(r) + 2\alpha_a r - 10 \log_{10} \left( \frac{(p_{et} g_0 c_w \Psi)^2}{32 \pi^2} \right) - 2S_a \text{corr}, \quad \text{and} \quad (4.1)
\]

\[
TS = P_{er} + 40 \log_{10}(r) + 2\alpha_a r - 10 \log_{10} \left( \frac{(p_{et} g_0 \lambda^2)^2}{16 \pi^2} \right) - 20 \log_{10} \left( \frac{g(\alpha, \beta)}{g_0} \right), \quad (4.2)
\]

where $r$ (m) is the range to the sphere, $P_{er}$ (dB re 1 W) is the received power, $p_{et}$ (W) is the transmit power, $\alpha_a$ (dB m$^{-1}$) is the absorption coefficient, $g_0$ (dimensionless) is the transducer on-axis (peak) gain, $g(\alpha, \beta)$ (dimensionless) is the transducer gain in the direction of off-axis angles $\alpha$ and $\beta$ (°), $\lambda$ (m) is the wavelength, $f$ (Hz) is the frequency, $c_w$ (m s$^{-1}$) is the water sound speed, $\tau$ (s) is the transmit pulse duration, $\Psi$ (sr) is the...
equivalent two-way beam angle, and $S_{a\text{corr}}$ (dB re 1) is the $S_a$ correction factor. Software (e.g. LSS by IMR in Bergen, Norway; or Echoview by Myriax in Hobart, Tasmania) can be used to convert raw data to echograms of $S_a$ or $TS$.

Although the ek5 data are less voluminous than the raw data, they do not contain the GPT settings (important metadata) and may contain quantization errors. For ER60 versions prior to 2.2.0, all versions of ES60, and possibly some or all versions of ES70, there is a quantization error in ek5 data that can cause significant (>4 dB) errors in the calibration results. This quantization error is extreme for measurements of highly reflective objects at a fixed range (e.g. a calibration sphere). Therefore, always record raw data and, unless you use ER60 version 2.2.0 or higher, always calibrate using raw data. If desired, ek5 data may be logged concurrently and independently.

Table 4.2. Key features of the EK60, ES60, and ES70 systems relevant to data outputs and calibration.

<table>
<thead>
<tr>
<th>Feature</th>
<th>EK60</th>
<th>ES60</th>
<th>ES70</th>
</tr>
</thead>
<tbody>
<tr>
<td>Primary purpose s</td>
<td>Scientific echosounder</td>
<td>Fishery echosounder</td>
<td>Fishery echosounder</td>
</tr>
<tr>
<td>Simrad supports as a scientific instrument</td>
<td>Yes</td>
<td>No</td>
<td>No</td>
</tr>
<tr>
<td>System type</td>
<td>Split-beam (except 12 kHz and custom units)</td>
<td>Split- and single-beam</td>
<td>Split- and single-beam</td>
</tr>
<tr>
<td>Outputs raw data</td>
<td>Yes</td>
<td>Yes (V1.4.3.64 +)</td>
<td>Yes</td>
</tr>
<tr>
<td>Outputs ek5 data</td>
<td>Yes, to file and Ethernet</td>
<td>Yes, to Ethernet; log with Echolog500 (<a href="http://www.echoview.com">www.echoview.com</a>); quantization error can affect calibrations, but may not affect sampling</td>
<td>Yes, to file and Ethernet. Uncertain if ek5 data have quantization error</td>
</tr>
<tr>
<td>Calibration tools</td>
<td>Yes</td>
<td>No</td>
<td>No</td>
</tr>
<tr>
<td>Triangle-wave error sequence (TWES; Section 4.1.1.2)</td>
<td>No</td>
<td>Yes</td>
<td>Yes</td>
</tr>
<tr>
<td>Record of GPT parameters</td>
<td>Calibration.exe updates trlist.ini</td>
<td>Manually update trlist.ini</td>
<td>Manually update trlist.ini</td>
</tr>
</tbody>
</table>

4.1.1.2 Triangle-wave error

The raw data from ES60 and ES70 echosounders are modulated with a triangle-wave error sequence (TWES) with a 1-dB peak-to-peak amplitude and a 2720-ping period (Figure 4.1) (Ryan and Kloser, 2004). The TWES averages to zero over a complete period, so its contribution to sampling error may be insignificant. However, the TWES
may bias calibration results as much as ±0.5 dB. Therefore, prior to calibrations, the TWES should be removed from ES60 or ES70 $P_{er}$ data.

To remove the TWES, it is first necessary to determine its phase. This is done by analyzing $P_{er}$ data collected during and shortly after the transmit pulse (the first few metres of the echogram) from at least 1360 continuous pings (half the period of the TWES). The TWES phase is defined relative to the first ping in the file. The $P_{er}$ data are corrected by subtracting the phase-aligned TWES from $P_{er}$ data from each transmission in each file. This procedure can be conveniently accomplished using the batch processing feature of ES60adjust.jar (an open source utility developed by CSIRO Marine and Atmospheric Research, available from the ICES website).

Figure 4.1. The ES60 and ES70 triangle wave error sequence (TWES) showing the 1-dB peak-to-peak amplitude and 2720-pings period. The expanded portion shows an underlying step function, due to sampling quantization, with 16-ping width and 1.176 x 10^{-2} dB height.

4.1.1.3 Settings

Nominal values for the GPT settings (Table 4.1) are listed in the TRList.ini file for each combination of transducer, transceiver, and pulse duration (Table 4.3). This file is located in the echosounder-software installation directory. TRList.ini can be updated through the ER60 menu or manually.

Table 4.3. Excerpt from a TRList.ini file showing nominal settings for a 38-kHz split-beam GPT, an ES38B transducer, and five pulse lengths (i.e. durations).

<table>
<thead>
<tr>
<th>GPTTRANSCEIVER=&quot;GPT-Q38(1)-F1&quot;</th>
</tr>
</thead>
<tbody>
<tr>
<td>BEAMTYPE=SPLIT</td>
</tr>
<tr>
<td>MARKETSEGMENT=FISHERY</td>
</tr>
<tr>
<td>MINMEDMAXTXPULSELENGTH=0.000256 0.000512 0.001024 0.002048 0.004096</td>
</tr>
<tr>
<td>ENDGPTTRANSCEIVER</td>
</tr>
<tr>
<td>TRANSUDER=&quot;ES38B&quot;</td>
</tr>
<tr>
<td>FREQUENCY=38000</td>
</tr>
<tr>
<td>BEAMTYPE=SPLIT</td>
</tr>
<tr>
<td>EQUIVALENT2WAYBEAMANGLE=-20.6</td>
</tr>
</tbody>
</table>
For each scientific echosounder transducer, Simrad measures the beamwidths, $\theta_{-3\,dB}'$ (°), and angle sensitivities, $A'$ (electrical ° / geometrical °), under controlled, freshwater tank conditions (Table 4.4). Simrad then uses the estimated $\alpha_{-3\,dB}'$ and $\beta_{-3\,dB}'$ values to estimate the equivalent two-way beam angle \[ \Psi' = 10 \log_{10}(\alpha_{-3\,dB}' \beta_{-3\,dB}' / 5800) \]. However, once the transducer is relocated to a different environment and mounted in a platform (e.g. the hull of a ship, centreboard, or towed body), these values may be different (Simmonds, 1984a, 1984b; Reynisson, 1998). This is because the local $c_w$ affects $\lambda$, $A$, $\theta_{-3\,dB}$, $\Psi$, and $g_0$, and, therefore, measures of $TS$ and $S_v$ (Bodholt, 2002):

\[
A = A'(c_w' / c_w) ;
\]

\[
\theta_{-3\,dB} = \theta_{-3\,dB}'(c_w / c_w') ;
\]

\[
\Psi = \Psi'(c_w^2 / c_w')^2 ; \text{ and}
\]

\[
g_0 = g_0'(c_w^2 / c_w')^2 .
\]

Measures of $A$ and $\theta_{-3\,dB}$ require echosounder-independent measures of the target location (Reynisson, 1990, 1998). Because such measures are difficult to make routinely (Simmonds, 1984a), use the nominal value for $A$, and derive $\psi$ from Equation (4.6) using a measure of local $c_w$, and $\Psi'$ from the manufacturer’s measurements at $c_w'$ (e.g. Table 4.4). Then, estimate values for $\alpha_{-3\,dB}$ and $\beta_{-3\,dB}$ from the calibration results. In this case, the estimates of $\alpha_{-3\,dB}$ and $\beta_{-3\,dB}$ may not indicate the true beamwidths, but their use in the directivity-pattern model will correctly compensate measures of $\sigma_{bs}$ for the directivity pattern (Bodholt, 2002).
Table 4.4. An example report of Simrad’s freshwater tank measurements for a Simrad ES38-B transducer.

<table>
<thead>
<tr>
<th>TRANSDUCTOR MEASUREMENTS</th>
<th>Part 1</th>
<th>Part 2</th>
<th>Part 3</th>
<th>Part 4</th>
<th>All Parts</th>
</tr>
</thead>
<tbody>
<tr>
<td>Measured at: f (kHz)</td>
<td>38.0</td>
<td>38.0</td>
<td>38.0</td>
<td>38.0</td>
<td>38.0</td>
</tr>
<tr>
<td>Beamwidth</td>
<td></td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>Longitudinal: B1 (deg)</td>
<td></td>
<td></td>
<td></td>
<td></td>
<td>6.8</td>
</tr>
<tr>
<td>Transversal: B2 (deg)</td>
<td></td>
<td></td>
<td></td>
<td></td>
<td>6.9</td>
</tr>
<tr>
<td>Directivity index (dB):</td>
<td></td>
<td></td>
<td></td>
<td></td>
<td>10 log(2.5/(sin(31/2)*sin(32/2))) 28.5</td>
</tr>
<tr>
<td>Equivalent two way</td>
<td></td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>beam angle (dB):</td>
<td></td>
<td></td>
<td></td>
<td></td>
<td>10 log(31*32/6300) -21.0</td>
</tr>
<tr>
<td>Impedance</td>
<td></td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>Z (ohm)</td>
<td>63.4</td>
<td>61.4</td>
<td>68.2</td>
<td>62.4</td>
<td>15.8</td>
</tr>
<tr>
<td>Phase (deg)</td>
<td>-2.7</td>
<td>-2.3</td>
<td>-6.0</td>
<td>0.8</td>
<td>1.5</td>
</tr>
<tr>
<td>Transmitting response</td>
<td></td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td></td>
<td>(at 1 metre)</td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>Sl (dB re 1 μPa/A):</td>
<td>208.6</td>
<td>208.7</td>
<td>209.5</td>
<td>208.7</td>
<td>209.1</td>
</tr>
<tr>
<td>Su (dB re 1 μPa/V):</td>
<td>172.6</td>
<td>172.0</td>
<td>173.1</td>
<td>172.8</td>
<td>185.1</td>
</tr>
<tr>
<td>Receiving sensitivity</td>
<td></td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td></td>
<td>Theoretical open circuit</td>
<td>-177.0</td>
<td>-177.0</td>
<td>-176.2</td>
<td>-177.0</td>
</tr>
<tr>
<td></td>
<td>(dB re 1 V/μPa)</td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>Leaking resistance (Mohm)</td>
<td>OK</td>
<td>OK</td>
<td>OK</td>
<td>OK</td>
<td>OK</td>
</tr>
<tr>
<td>Type:</td>
<td>ES38-B</td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>Serial no:</td>
<td>30596</td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>Watertemp:</td>
<td>21.0 °C</td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
<tr>
<td>Tested by:</td>
<td>BK</td>
<td></td>
<td></td>
<td></td>
<td></td>
</tr>
</tbody>
</table>

4.1.1.4 Preparation

Before a calibration, Simrad (2001) recommends checking the internal test signal for each frequency (channel) to identify any major operational issues with the echosounder receiver. If the internal test signal has not been disconnected by the manufacturer, with the transceiver channel set to “Test” mode, record the Noise Est(imate) displayed in the “Numerical View”. Acceptable test values are −70 ± 1 or −64 ± 1 dB re 1 W for a single- or split-beam system, respectively.

Record the ambient noise at each frequency to compare with conditions at historical or future calibration sites. With the transceiver channel set to “Passive” mode, record the range of background noise values estimated under “Environment” in the “Numerical View” (ER60 software only).

Deploy a conductivity–temperature–depth probe (CTD) at the calibration site to obtain depth profiles of salinity and temperature. Calculate the mean salinity and temperature, and harmonic mean sound speed (see Section 2.1.2.5) for the range between the transducer and the calibration sphere. In the “ER60 Install-Environment” dialog box, input these values and calculate absorption coefficients for each frequency (Francois and Garrison, 1982a). For ES60 and ES70 echosounders, input the harmonic mean sound speed (see Section 2.1.2.5) calculated from the measured salinity and temperature and indicate if the environment is fresh- or salt water. Note, these values may be set or changed when later processing the raw data, e.g. with Echoview (Myriax, 2014).

The effects of non-linear harmonic distortion can affect the transducer directivity pattern and limit the sound energy radiated into the water (Tichy et al., 2003; Korneliussen...
et al., 2008). These non-linear effects generally increase with frequency. To mitigate these effects, (Korneliussen et al., 2008) recommended maximum $p_{et}$ for several Simrad transducers (Table 4.5).

Table 4.5. Recommended maximum input power ($p_{et}$) for 60% electro-acoustic efficiency, for some common transducer sizes (Korneliussen et al., 2008).

<table>
<thead>
<tr>
<th>Frequency (Hz)</th>
<th>18</th>
<th>38</th>
<th>70</th>
<th>70</th>
<th>120</th>
<th>200</th>
<th>333</th>
<th>400</th>
</tr>
</thead>
<tbody>
<tr>
<td>Transducer area ($10^{-3}$ m$^2$)</td>
<td>200</td>
<td>100</td>
<td>30</td>
<td>12</td>
<td>10</td>
<td>4.4</td>
<td>1.6</td>
<td>1.1</td>
</tr>
<tr>
<td>Nominal $\theta_{-3 , \text{dB}}$ (°)</td>
<td>11</td>
<td>7</td>
<td>7</td>
<td>11</td>
<td>7</td>
<td>7</td>
<td>7</td>
<td>7</td>
</tr>
<tr>
<td>Maximum $p_{et}$ (W)</td>
<td>5 000</td>
<td>2 500</td>
<td>750</td>
<td>300</td>
<td>250</td>
<td>110</td>
<td>40</td>
<td>28</td>
</tr>
</tbody>
</table>

Generally, echosounders should be calibrated in the configuration that they will be used for sampling. If multiple echosounders with different frequencies will be used during sampling, those echosounders should also be active during calibration. To reduce the time necessary to calibrate, the transmit rate may be maximized, generally without deleterious effects. However, if the transmit rate is too rapid, echoes from a previous ping might alias into data from the current transmission, introducing “phantom seabed” or “false bottom” echoes in the echograms for one or more frequencies. If this occurs, adjust the transmit rate until the aliased reverberation is avoided.

Generally, when calibrating, a single sphere is suspended below the transducer. If multiple spheres are suspended concurrently (e.g. to efficiently calibrate multiple systems), MacLennan (2011) cautions that forward scatter may introduce some measurement error.

### 4.1.2 Split-beam EK60

This section describes three approaches to calibrating a split-beam EK60 with a sphere including: (i) the recommended ER60 Calibration.exe utility, (ii) an alternative ExCal script, and (iii) an on-axis method. The differences, advantages, and limitations of each approach are discussed.

#### 4.1.2.1 Calibration.exe

Measurements of $TS$ from a sphere are made as it is moved throughout the transducer beam. Calibration.exe uses these measures to estimate $\alpha_{-3 \, \text{dB}}$, $\beta_{-3 \, \text{dB}}$, $\alpha_{o}$, $\beta_{o}$, $G_{0}$, and $S_{a \, \text{corr}}$. The steps are detailed below and in Simrad (2008).

#### 4.1.2.1.1 Data collection

Calibration.exe requires measurements of $TS$ from a sphere throughout all the transducer-beam quadrants. The apparatus and procedure to map the directivity pattern are described in Chapter 2.

To measure the sphere $TS$, it generally suffices to use the default parameters for single-target detection (i.e. minimum echo length = 0.8, maximum echo length = 1.8, maximum phase deviation = 8.0, maximum gain compensation = 6.0, minimum echo spacing = 1.0). However, the minimum $TS$ threshold should be set $-10$ dB below the theoretical $TS$ value ($TS_{\text{theory}}$) for the reference sphere value. For example, for a sphere with $TS_{\text{theory}} = -42.2$ dB, set the minimum $TS$ threshold to $-52$ dB. If it is difficult to locate the sphere, lower the minimum $TS$ threshold, find the sphere, and then restore the threshold to the recommended level.
Launch Calibration.exe from the ER60 "Single Target Detection" dialogue box. Input the $T_S^\text{theory}$ (Chapter 2), set the allowable $T_S$ deviation = 5 dB, and set the upper and lower ranges equal to the on-axis sphere range ±3 m. These settings allow most of the sphere echoes to be recorded. Spurious echoes from other scatterers (e.g. fish, swivels, and weights) may be filtered during data processing. To assure that the transducer beam is sufficiently mapped, Calibration.exe displays the positions of the $T_S$ measurements. To precisely estimate the $a_{-3\,\text{dB}}, \beta_{-3\,\text{dB}}, a_\varphi, \beta_\varphi, G_0$, and $S_{a\,\text{corr}}$ values, at least 100 $T_S$ detections should be evenly distributed throughout the $a_{-3\,\text{dB}}$ and $\beta_{-3\,\text{dB}}$ (e.g. to ±3.5° for each axis of a 7°-beamwidth transducer), and ten or more detections should be near the beam axis (Simrad, 2008).

If the transducers to be calibrated are mounted sufficiently close to each other or the ranges to the sphere are sufficiently large, or both, $T_S$ data can be collected simultaneously at multiple frequencies (Demer et al., 1999; Conti et al., 2005a; Korneliussen et al., 2008). Then, each channel may be calibrated by replaying the same raw data file and rerunning Calibration.exe. For example, data used to calibrate 38-, 70-, 120-, and 200-kHz echosounders may be collected at the same time by moving a 38.1-mm-diameter sphere made from tungsten carbide with 6% cobalt throughout the overlapping beams. In this case, the Calibration.exe should be first configured for the most centrally located transducer, and the $T_S$ data should be collected to the edges of its beam. Then, in succession, the raw data file(s) can be repeatedly replayed, each time configuring Calibration.exe for each one of the other transducers. If one or more of the beams are insufficiently mapped, collect more $T_S$ data and replay again using both raw-data files.

### 4.1.2.1.2 Data processing

After sufficient $T_S$ data are collected, save the file, fit the beam model, and inspect and edit the results. The beam model has a theoretical basis and, with a properly functioning system, is less susceptible to noise than the polynomial model. However, if the system is dysfunctional, the polynomial model can provide additional diagnostic information (Figure 4.2).

Calibration.exe displays the $T_S$ data overlaid on the selected beam model. Optional views are polar or one of several planar slices. $T_S$ outliers can be deselected with a click of the mouse cursor, and the beam model can be refitted. The root-mean-square (rms) error generally increases with frequency. An rms error <0.4 indicates an acceptable model fit. Fits with larger rms values should be scrutinized and perhaps discarded. Examinations of the polar and planar views can confirm that the system is working properly. For example, the plot view of the beam model for a circularly symmetrical transducer should display concentric circles. If it displays ellipsoids with the major axis at 45° or 135°, the transducer or receiver might have a dysfunctional quadrant (e.g. Figure 4.2). If non-circular shapes are paired with large offset angles, the transducer may be wired incorrectly. See Simrad (2008) for additional troubleshooting guidance.

If the estimated $G_0$ differs from that of recent calibrations conducted under similar environmental conditions by >0.5 dB (ICES, 1994), the system should be evaluated for proper function, the calibration should be repeated, or both.
Figure 4.2. Example beam maps for a transducer with a disconnected upper-right quadrant. In this case, $TS$ measurements could not be made towards the edges of the upper-right and lower-left quadrants. The beam model (left) has concentric lines, but banding in the positive (red) and negative residuals (blue) indicates a problem. The polynomial model (right) shows slanted ellipsoids with more randomly distributed residuals, also indicating a problem with the transducer or receiver.

4.1.2.1.3 Echosounder update

The EK60 can be configured with the updated calibration parameters by selecting “Update Beam Data” in Calibration.exe. Optionally, record the calibration parameters manually and exit Calibration.exe without updating the beam data; then set the transducer parameters through “Install Transducer Parameters” in the ER60. Both of these methods update the calibration values in the TRlist.ini file (Section 4.1.1.3). A third option is to make a backup copy of Trlist.ini (e.g. named Trlist.ini.bak) and then manually update the TRlist.ini file using a text editor. If desired, a simplified version of Trlist.ini file may be created by deleting all of the >5000 lines of text that are not relevant to the system being calibrated. A fourth option is to simply retain the current values in the echosounder and then calibrate the raw data while processing, e.g. using an .ecs file in Echoview (Myriax, 2014). With this latter option, however, the EK60 will not display calibrated data.

4.1.2.1.4 Advantages and limitations

The following are advantages of Calibration.exe:

- a simple application designed for EK60 calibrations;
- runs within the ER60 in normal and replay modes;
- runs outside the ER60 for analysing calibration data sets;
- provides model-fit rms error to judge the quality of calibrations; and
- readily configures the EK60 with the calibration results.

The following are limitations of Calibration.exe:

- records only the first $TS$ value measured in each cell of a 25 x 25 array spanning the –6 dB points of the beam;
- the beam-model is fitted with equally weighted $TS$ measurements, perhaps resulting in a fit that exceeds all of the near-axis $TS$ measurements;
Calibration of acoustic instruments

- $T_S$ measurements must span the beam, which may be time-consuming or unachievable in poor conditions or when the sphere movement is otherwise physically constrained; and
- noisy or biased $T_S$ data (e.g. echoes from the sphere among biota) cannot be readily filtered and might bias the calibration results.

4.1.2.2 ExCal
ExCal is a free open-source Matlab (The Mathworks, MA, USA) script for estimating calibration parameters for EK60, ES60, and ES70 echosounders. It reads raw data files.

4.1.2.2.1 Advantages and limitations
The following are advantages of the ExCal utility:
- offers convenient exclusion of all undesired $T_S$ data;
- uses more $T_S$ data than Calibration.exe to calculate $G_0$ and $S_a$ corr;
- provides options to filter $T_S$ and thereby allow better utilization of noisy data;
- estimates $G_0$ and $S_a$ corr using maximum, mean, or model-fit $T_S$ values;
- plots the calibration data and results for use in calibration reports; and
- is open source code that can be modified to test other algorithms.

The following are limitations of the ExCal utility:
- does not update the EK60 with new calibration parameters; and
- requires a Matlab license and familiarity with Matlab.

4.1.2.3 On-axis calibration
A third approach to estimating $G_0$ and $S_a$ corr is to make measurements of $T_S$ and $S_a$ with the sphere positioned on the beam axis. This is most accurately accomplished using split-beam angle measurements to position the sphere and a post-processing application, e.g. Echoview (Myriax, 2014), to filter the measurements and calculate mean $S_a$. To position the sphere on the axis of a single-beam transducer, it is necessary to iteratively maximize the measured $T_S$ (Section 4.1.4).

4.1.2.3.1 Data collection
To position the sphere on the beam axis, surround the sphere echo with a 5-m wide active layer in the ER60 “Echogram” view and monitor the location of the sphere in the beam using the ER60 “Single Target Position” view. This procedure restricts the targets appearing in the “Single Target Position” view to only those within the layer that meet the “Single Target Detection” criteria (Section 4.1.2.1). Calculate the mean of approximately 100 sphere-$T_S$ measurements and compare it to $T_S$theory. Using an appropriate sphere for each frequency, repeat this process for each transducer. Collect separate raw data files for each on-axis calibration.

4.1.2.3.2 Data processing
Using an application to process the raw data, e.g. Echoview (Myriax, 2014), adjust the “Single Target Detection” criteria (Table 4.6) to detect sphere echoes. Filter echoes that are not from the sphere and restrict the sphere-$T_S$ measurements to those exactly or approximately on the beam axis.
Table 4.6. Recommended “Single Target Detection” criteria for Echoview (Myriax, 2014) processing of TS measurements of a 38.1-mm-diameter WC sphere at 38 kHz. These values might require adjustments for different spheres, frequencies, and environmental conditions.

<table>
<thead>
<tr>
<th>“Single Target Detection” criteria</th>
<th>Recommended setting</th>
</tr>
</thead>
<tbody>
<tr>
<td>TS threshold</td>
<td>-50.0 dB</td>
</tr>
<tr>
<td>Pulse length determination level</td>
<td>6.0</td>
</tr>
<tr>
<td>Minimum normalized pulse length</td>
<td>0.20*</td>
</tr>
<tr>
<td>Maximum normalized pulse length</td>
<td>1.50</td>
</tr>
<tr>
<td>Beam compensation model</td>
<td>Simrad LOBE</td>
</tr>
<tr>
<td>Maximum beam compensation</td>
<td>3.0 dB</td>
</tr>
<tr>
<td>Maximum standard deviation of minor axis angles</td>
<td>0.60°</td>
</tr>
<tr>
<td>Maximum standard deviation of major axis angles</td>
<td>0.60°</td>
</tr>
<tr>
<td>Filter beam compensation</td>
<td>0.1 dB**</td>
</tr>
</tbody>
</table>

*Less than the ER60 default value.
**Increase in noisy conditions.

Calculate the mean on-axis sphere TS by averaging in the linear domain. Estimate the new $G_0$ using the following equation.

$$\text{new } G_0 = \frac{TS - TS_{\text{theory}}}{2} + \text{old } G_0 \tag{4.8}$$

Calculate the theoretical on-axis $S_a (S_{a \text{ theory}})$ for the sphere (see Chapter 2). Use the mean measured sphere $S_a$ and the following equation to estimate the new $S_{a \text{ corr}}$.

$$\text{new } S_{a \text{ corr}} = (\text{old } G_0 + \text{old } S_{a \text{ corr}}) - 10 \log \left( \frac{S_{a \text{ theory}}}{S_a} \right) - \text{new } G_0 \tag{4.9}$$

Update the echosounder settings using any of the manual options described in Section 4.1.2.1.

### 4.1.2.3.3 Advantages and limitations

The following are advantages of on-axis calibrations:

- data can be filtered using raw data-processing software, e.g. Echoview (Myriax, 2014);
- calibration is quicker and more successful when conditions or apparatus prevent beam mapping; and
- $S_{a \text{ corr}}$ is estimated using more data than used in Calibration.exe.

The following are limitations of on-axis calibrations:

- requires that the sphere is centred stably on the beam axis;
- uses existing $\theta_{-3 \text{ dB}}, \theta_o,$ and $A$ settings to calculate the sphere position;
- does not estimate beam shape, $\theta_{-3 \text{ dB}}$ or $\theta_o$ values;
- does not automatically update the ER60 with the new calibration settings; and
- does not verify the function of all quadrants in a split-beam system (e.g. Figure 4.2).
4.1.3 Split-beam ES60 and ES70

ES60 and ES70 software packages do not include a calibration utility. Nevertheless, these echosounders may be calibrated. This section outlines the calibration procedures that are unique to ES60 and ES70 echosounders.

4.1.3.1 Data collection

The apparatus for calibrating and a method for positioning the sphere are described in Chapter 2. Calibrate the echosounders using the same \( p_{et} \) and \( \tau \) to be used for acoustic sampling. Assure that \( p_{et} \) does not exceed the values in Table 4.5.

Position and monitor the sphere within the acoustic beam using the “Echo Trace” display in the ES60 or the “Fish Position” display in the ES70. To display only the sphere detections, adjust the start and stop ranges in the “Echogram” display. Check that adjustments to the sphere position are mimicked in the echosounder measurements. Methodically move the sphere and record its \( T_{S} \) uniformly throughout the beam. If software is not available to track the sphere movements, either note on a piece of paper where the sphere has been or tape clear plastic over the “Echo Trace” or “Fish Position” display and mark the sphere locations with a pen. For each calibration, start a new file and assure that the raw data includes more than 1360 pings so that the TWES (Section 4.1.1.2) can be removed prior to data processing.

4.1.3.2 Data processing

ES60 and ES70 calibration data can be processed using one of the methods described for EK60 calibrations (Section 4.1.2). To use Calibration.exe or the on-axis approach, first generate new raw files with the TWES removed (Section 4.1.1.2). Uncorrected raw files can be processed directly with ExCal, as it can remove the TWES.

4.1.3.3 Echosounder update

The calibration settings in split-beam ES60 and ES70 echosounders can be updated by manually editing the TRlisit.ini file (Section 4.1.2.1). Alternatively, the uncalibrated raw data may be calibrated using processing software, e.g. Echoview (Myriax, 2014).

4.1.3.4 Advantages and limitations

The advantages and limitations of ES60 and ES70 calibrations are the same as for the three approaches to EK60 calibrations (Section 4.1.2), except that the update of calibration settings is not automated.

4.1.4 Single-beam ES60 and ES70

Single-beam echosounders do not measure the angular positions of the sphere within the transducer beam. Therefore, to position the sphere on the beam axis, the sphere is moved until the beam-uncompensated \( T_{S} \) is maximized (Chapter 2). Values for \( G_{a} \) and \( S_{a,corr} \) are then estimated (see Section 1.2.2).

4.1.4.1 Data collection

Adjust the display range and colour scale to highlight the sphere echo in the ES60 or ES70 Echogram window. The colour-scale resolution is only 3 dB, but other software applications, e.g. Echoview Live Viewing (Myriax, 2014) can provide better display resolution. Start a new file and assure that the raw data from the calibration include more than 1360 pings so that the TWES (Section 4.1.1.2) can be removed prior to data processing.
4.1.4.2 Data processing

First remove the TW ES from the raw data (Section 4.1.1.2). Then, apply a single-target detection algorithm to the corrected data and, using processing software, e.g. Echoview (Myriax, 2014), export the sphere-TS measurements in csv-format file. Using these data, estimate the mean on-axis sphere-TS by averaging the largest 5% of the TS measurements. Estimate the new $G_0$ (Section 4.1.2.3). Next, calculate the mean on-axis $S_a$ using the sphere-$S_p$ measurements that correspond to the largest 5% of the TS measurements. Calculate the new $S_a_{corr}$ (Section 4.1.2.3.2).

4.1.4.3 Echosounder update

The calibration settings in single-beam ES60 and ES70 echosounders can be updated by manually editing the TRlisit.ini file (Section 4.1.2.1). Alternatively, the uncalibrated raw data may be calibrated using processing software, e.g. Echoview (Myriax, 2014).

4.1.4.4 Advantages and limitations of ES60 and ES70 single-beam calibrations

The advantages and limitations of single-beam calibrations are the same as for on-axis calibrations (Section 4.1.2.3.2).

4.1.5 Measurement error

Calibration uncertainty is presented in Chapter 3. Details specific to EK60, ES60, and ES70 systems are discussed here. The chosen approach (e.g. Calibration.exe, ExCal, or on-axis) can result in different estimates of $G_0$ and $S_{a_{corr}}$. In most instances, however, the various estimates of $G_0$ vary by approximately ±0.1 dB. One reason for small discrepancies is that the fitted beam model often exceeds slightly the measurements near the beam axis (e.g. Figure 4.3).

![Figure 4.3](image)

Figure 4.3. The beam-model fit to sphere-TS measurements made with an EK60 configured with an ES38-B transducer. In this example, the TS measurements near the beam axis are all below the model.

Knudsen (2009) estimated the precision of sphere-calibration estimates of $G_0$ to be ±0.1 dB for an EK60 configured with a Simrad ES38-B transducer (Figure 4.4). Calibrated $G_0$ may be a little less precise for higher frequencies, e.g. 70, 120, and 200 kHz.
The aforementioned estimates of precision for $G_0$ were for contemporary calibrations made in low-noise conditions (i.e. low sea state and resolvable sphere echoes). Demer et al. (1999) showed that the precision of split-beam angle measurements decreases with the signal-to-noise ratio. Accordingly, Pedersen (2007) found that the standard error in some beam-model parameters (i.e. $\alpha_{-3\ dB}$, $\beta_{-3\ dB}$, $\alpha_o$, and $\beta_o$) increased with ambient noise level. Changes in temperature also affect transducer performance (Demen and Renfree, 2008). Of the 10 Simrad transducers they tested, the composite transducers were less sensitive to changes in temperature than the older models, and the ES120-7 was most sensitive. Transducer performance is also affected by changes in pressure (Chapter 2). For example, Dalen et al. (2003) observed a 2 dB change in sphere TS when a Simrad ES38DD transducer was lowered from near the surface to 500 m. It is, therefore, important to calibrate echosounder systems under the range of conditions (i.e. temperature, salinity, and pressure) to be encountered during the acoustic sampling.

### 4.1.6 Documentation

Calibration metadata provide a record that is useful for quality control and comparisons of the results. These metadata (Anon., 2014) can be organized by physical environment (site and water column), equipment and settings, and calibration results (Table 4.7).

Table 4.7. Recommended calibration metadata.

<table>
<thead>
<tr>
<th>Site</th>
<th>Platform name, location, latitude, longitude, seabed depth</th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
<td>Sea-state at start and end of exercise</td>
</tr>
<tr>
<td></td>
<td>Date and time at start and end of calibration, with time zone</td>
</tr>
<tr>
<td></td>
<td>Comments (e.g. anchoring notes)</td>
</tr>
<tr>
<td>Water column</td>
<td>CTD data including model, configuration, and filename</td>
</tr>
</tbody>
</table>
Water temperature, salinity, and sound speed at the transducer and sphere depths

Harmonic-mean sound speed between the transducer and the sphere

Mean absorption coefficient between transducer and sphere, for each frequency

Estimated noise, for each frequency

Comments (e.g., other scatterers present)

**Equipment**

- Transducer model, frequency, serial number, and depth
- GPT model, frequency, serial number, and firmware version
- Echosounder software version
- Sphere identification, material, diameter, range from transducer, and \( T_{S_{\text{theory}}} \)

**Settings (by frequency)**

- Transmit power, \( p_{et} \) (W)
- Pulse length (duration), \( \tau \) (ms)
- On-axis gain, \( G_0 \) (dB)
- \( S_a \) correction, \( S_a\_\text{corr} \) (dB)
- Bandwidth, \( b_f \) (Hz)
- Sample interval (m)
- Equivalent two-way beam angle, \( \Psi \) (dB re 1 sr)
- Absorption coefficient, \( a_a \) (dB/km)
- Water sound speed, \( c_w \) (m/s)
- Angle sensitivity alongships, \( \Lambda_a \) (electrical°/geometric°)
- Angle sensitivity athwartships, \( \Lambda_{\beta} \) (electrical°/geometric°)
- 3-dB beamwidth alongships, \( a_{-3\,\text{dB}} \) (°)
- 3-dB beamwidth athwartships, \( \beta_{-3\,\text{dB}} \) (°)
- Angle offset alongships, \( \alpha_o \) (°)
- Angle offset athwartships, \( \beta_o \) (°)
- Single-target detection parameters

**Results**

- Internal test oscillator readings
- On-axis data-collection start and end times
- Beam map start and end times
- Processing software name and version
- On-axis results including mean \( T_S \) and mean \( S_a \)
- Beam model results including estimated \( a_{-3\,\text{dB}} \), \( \beta_{-3\,\text{dB}} \), \( a_o \), \( \beta_o \), \( G_0 \), and \( S_a\_\text{corr} \) and model fit rms error

### 4.1.7 Quick start

The following is a guide to using ER60 and its Calibration.exe utility to calibrate EK60 split-beam systems. Similarly, ES60 or ES70 split-beam systems may be calibrated by replaying TWES-corrected raw data (Section 4.1.1.2) in ER60 and Calibration.exe.

1) Record metadata (Section 4.1.6) throughout the calibration.

2) Anchor in sheltered water with few biota (there may be fewer during daytime).

3) Set the “Raw Data Collection Range” to include the seabed. Ensure that the “Display Range” and “Bottom Detection Range” do not exceed “Data Collection Range”.


4) Maximize the ping rate.

5) Set the ER60 “Single-Target Detection” parameters.

6) Install the calibration apparatus and position the sphere at a range two- to three-fold the transducer far-field range, e.g., 15–25 m below the transducer, on or near its beam axis.

7) Measure the seawater temperature and salinity vs. depth. Use these data to estimate the water sound speed, \( c_w \) (m s\(^{-1}\)), values at the depth of the transducer and the sphere, and the harmonic-mean value between them. Also estimate the mean absorption coefficient, \( \alpha_a \) (dB m\(^{-1}\)), over the same depth range for each frequency, \( f \) (Hz). Update these environmental settings in the GPT.

8) Calculate \( T^S_{theory} \) using the \( c_w \) estimated for the depth of the sphere.

9) Launch Calibration.exe from the ER60 “Single Target Detection” dialogue box. Input \( T^S_{theory} \), the allowable \( T^S \) deviation = 5 dB, and the upper and lower measurement ranges = sphere range ±3 m).

10) Record at least 100 sphere-\( T^S \) measurements as it is moved throughout the beam to at least the edges of the nominal half-power beamwidth.

11) Record at least 10 sphere-\( T^S \) measurements near the beam axis.

12) Stop recording and save the file. Examine the sphere-\( T^S \) measurements overlaid on the beam-model fit, in both polar and planar views. Remove any outliers and re-save the file.

13) Confirm that the rms error for the beam-model is < ~0.4 (this value generally increases with frequency), and that the beam-model estimates of \( G_0 \) and \( S_{a,corr} \) are within ±0.5 dB of their expected values.

14) If desired, update the echosounder settings manually or, if using Calibration.exe, using “Update Beam Data”.

### 4.2 Simrad ME70

#### 4.2.1 Introduction

The Simrad ME70 multibeam echosounder transducer is composed of 800 elements configured in a 20 x 40 matrix. The interelement spacing is 15 mm alongships and 7.5 mm athwartships, forming a 30-cm\(^2\) transducer surface. Operating in fishery mode (Cutter et al., 2010), the ME70 can be configured to produce up to 45 steered, single- or split-beams (see Chapter 1), each spanning a portion of the total system bandwidth, 70–120 kHz (Trenkel et al., 2008). This section presents some theory and then calibration procedures for calibrating a vessel-mounted ME70 configured for operation with multiple split-beams operating at different centre frequencies. The procedure is based on the sphere method (see Section 4), applied to multiple beams (Ona et al., 2009).

#### 4.2.2 Theory

Relative to the other echosounders described in Section 4.1, the ME70 has some unique parameters and considerations. This section describes some features of the ME70 that are relevant to sphere calibration and data processing.
4.2.2.1 Adjust parameters

The ME70 software includes nominal values for on-axis gain, $G_0\text{\scriptsize nom}$ (dB re 1). The gain adjustment, $\Delta G_0$ (dB re 1) is the difference between $G_0\text{\scriptsize nom}$ and the calibrated gain, $G_0$ (dB re 1):

$$G_0 = G_0\text{\scriptsize nom} + \Delta G_0.$$  

(4.10)

Similarly, the ME70 software includes nominal values for $S_a\text{\scriptsize corr}$ correction, $S_a\text{\scriptsize corr\scriptsize nom}$ (dB re 1). The $S_a$ correction adjustment, $\Delta S_a\text{\scriptsize corr}$ (dB re 1) is the difference between $S_a\text{\scriptsize corr\scriptsize nom}$ and the calibrated $S_a$ correction, $S_a\text{\scriptsize corr}$ (dB re 1):

$$S_a\text{\scriptsize corr} = S_a\text{\scriptsize corr\scriptsize nom} + \Delta S_a\text{\scriptsize corr}.$$  

(4.11)

4.2.2.2 Bearing angles

The ME70 provides target bearing angles in the alongships, $\alpha$ (°) and athwartships, $\beta$ (°) planes for each echo sample within each beam. Using far-field beamforming, the electrical phase angle $\phi$ and target mechanical angle are related by:

$$\phi = -\frac{\pi}{\lambda} [ \sin(\theta_s) \theta_t - \sin(\theta_s) \theta_s ] ,$$  

(4.12)

where $\theta_t$ is the target mechanical angle (i.e. $\alpha$ or $\beta$), $\theta_s$ is the beam steering angle, and $d_{\text{eff}}$ is the effective distance between the acoustic centres of the split-beam subarrays. In the case where $\theta_t$ is close to $\theta_s$, Equation (4.12) can be approximated by:

$$\phi = \frac{2 \pi d_{\text{eff}} \cos(\theta_s)}{\lambda} (\theta_t - \theta_s) .$$  

(4.13)

The error in the ME70 angle conversion factor, $\lambda / [2 \pi d_{\text{eff}} \cos(\theta_s)]$, increases with increasing off-axis angle and beam steering angle (Figure 4.5).

![Figure 4.5. Differences between the target angles, $\theta_t$, calculated from Equation (4.13) and the true target angles, relative to the beam axis.](image)

The correct off-axis angle, $\theta_{t\text{corr}}$, can be obtained using:

$$\theta_{t\text{corr}} = \sin^{-1} \{ \cos(\theta_s)(\theta_t - \theta_s) + \sin(\theta_s) \} .$$  

(4.14)

To compute $\theta_{t\text{corr}}$ from an electrical angle in units of phase steps, as from the ME70 .raw CON0 datagram, use:
\[ \theta_{t \text{ corr}} = \sin^{-1} \left[ \frac{\cos(\theta_s)}{\cos(\alpha)} \frac{\pi}{128} \phi_{\text{step}} + \sin(\theta_s) \right], \]

(4.15)

where \( \phi_{\text{step}} \) ranges from –127 to 128 phase steps, and the angle sensitivity, \( \alpha = 180/(0.75 \theta_{-3 \text{ dB}}) \), is from the ME70 .raw file for each beam, in alongships or athwartships plane. In this case, \( \theta_{-3 \text{ dB}} \) is the –3 dB one-way beamwidth.

### 4.2.2.3 Beam axes

The ME70 beams are steered through classical beamforming. The signal phases for each of the transducer elements are aligned to transmit and receive coherently in the steered direction. For steered beams, the beam axis, the axis of maximum sensitivity, differs from the steering angle, the axis of zero phase.

The beam directivity pattern, \( b(\alpha, \beta) \), results from the product of the single-element directivity pattern, \( b_{el}(\alpha, \beta) \), and the beamformed-array directivity pattern, \( b_{array}(\alpha, \beta) \):

\[ b(\alpha, \beta) = b_{el}(\alpha, \beta)b_{array}(\alpha, \beta). \]

(4.16)

For small steering angles, \( b_{el}(\alpha, \beta) \) is much broader than \( b_{array}(\alpha, \beta) \), and effectively constant within \( b_{array}(\alpha, \beta) \). Therefore, the shape of \( b(\alpha, \beta) \) closely approximates \( b_{array}(\alpha, \beta) \), which is maximum for the steering angle. For large steering angles, however, \( b_{el}(\alpha, \beta) \) decreases rapidly enough to make the maximum \( b(\alpha, \beta) \) shift towards the maximum \( b_{el}(\alpha, \beta) \). This effect is taken into account in the directivity-pattern models used in the ME70 TS computations and in the directivity-pattern models implemented in Calibration.exe V. 1.2.5.

### 4.2.2.4 Motion compensation

ME70 steering angles may be compensated for vessel pitch and roll angles to optimize the accuracies of the range and bearing measurements. However, this motion compensation reduces the accuracies of the calibrated measurements of target strength, \( TS \) (dB re 1 m²) and volume backscattering strength, \( S_v \) (dB re 1 m² m⁻³) (figure 4.6). When the steering angles are compensated for motion, \( b(\alpha, \beta) \) and, therefore, \( G_0 \) change. To a first-order approximation, the beamwidths change by:

\[ \alpha_{-3 \text{ dB}} = \frac{\cos(\alpha + \text{pitch})}{\cos(\alpha)}, \text{ and} \]

(4.17a)

\[ \beta_{-3 \text{ dB}} = \frac{\cos(\beta + \text{roll})}{\cos(\beta)}, \]

(4.17b)

and both the on-axis directivity and the gain value change by the following factor:

\[ \frac{b(\alpha + \text{pitch}, \beta + \text{roll})}{b(\alpha, \beta)} \]

(4.18)

Although these effects are negligible for small steering angles, motion compensation should not be used during calibration and for subsequent measures of \( TS \) and \( S_v \). If motion compensation is used, then account should be made for the additional uncertainty in the measurements made in the most steered beams.
4.2.3 Calibration

The ME70 software includes a programme to calibrate configurations for measuring $T_S$ and $S_v$. The ME70 calibration utility is largely derived from the EK60 Calibration.exe utility. Therefore, it is assumed that readers of this section are familiar with EK60 calibration theory and procedures (see Chapters 1–3 and Section 4.1) and the ME70 manual (Simrad, 2012). The ME70 also includes some diagnostic utilities that should be run before a sphere calibration.

4.2.3.1 Diagnostics

In the ME70, 25 TRX32 cards control the signal transmission and reception on the 800 transducer elements (Figure 4.7). The ME70 software includes (i) BITE to test the function of each TRX32 card and each element and (ii) another utility to measure ambient acoustic noise within the bandwidths of the receivers.

![Figure 4.6. Example effect of roll correction (up to $1^\circ$) on the $T_S$ measurements of a sphere in a beam steered $57^\circ$. Biased detections are shown for five nominal sphere positions (blue). Overlaid are the measured $T_S$ at athwartships angles corrected using Equation (4.18) (red).](image)

![Figure 4.7. Each of 25 TRX32 cards, numbered 0–24 in the 5 x 5 matrix, control 32 of the 800 transducer elements.](image)

4.2.3.1.1 BITE test

The BITE test identifies and disables any dysfunctional elements. The test should be run before a sphere calibration because disabled elements or other system malfunction may affect the calibration results.
1) In the “SetUp” menu, select “BITE”.

The “B-Scan” tab presents the signal amplitude vs. time for the 800 transducer elements. Amplitudes during transmission are red and those during reception are blue.

2) To start the measurement, in the “Mode” field select “Element amplitude”.
3) To freeze the display, toggle “None/Freeze”.
4) To set the first and last samples to be displayed, fill the “Start Sample” and “Sample Range” fields. Using the 62.5 kHz sampling rate, the sample index can be interpreted as sample range (Table 4.8).

Table 4.8. Conversion of sample index to range.

<table>
<thead>
<tr>
<th>Sample Index</th>
<th>0</th>
<th>100</th>
<th>200</th>
<th>300</th>
<th>500</th>
<th>1,000</th>
<th>5,000</th>
</tr>
</thead>
<tbody>
<tr>
<td>Range (m)</td>
<td>0</td>
<td>1.2</td>
<td>2.4</td>
<td>3.6</td>
<td>6</td>
<td>12</td>
<td>60</td>
</tr>
</tbody>
</table>

5) In the “Decimation Method” field, select “Single”.
6) In the “Scale” field, select “Automatic”. If necessary, in the “Sample pixel size” field, change the (default) “Width” (1) and “Height” (6) values.

The “Matrix” tab presents the average amplitude of each element. If the averaging period includes a transmit pulse, the Matrix displays the beamforming amplitudes applied to each element.

7) To start the measurement, in the “Mode” field select “Element amplitude”.
8) To freeze the display, select “None/Freeze”.
9) To set the first and last samples of the analysis period, fill the “Start Averaging” and “Averaging Range” fields. Values for analysing either the transmission or reception can be determined from the y-axis of the B-scan view.
10) In the “Scale” field, select “Manual” mode and set “Minimum” to 35 dB and “Maximum” to 45 dB.
11) In the “Element size in pixel” field, set the “Width” and “Height” to have a 1:2 proportion (e.g. 12 and 24, respectively) matching the aspect of the transducer (i.e. 7.5 mm athwartships and 15 mm alongships).
12) In the “Transmit pulse” field, select “Current”.

Malfunctioning elements have weak, strong, or unstable levels during transmission, reception, or both (Figure 4.8). A large number of contiguous malfunctioning elements may be due to a faulty cable or power supply. However, element malfunctions are most often caused by a faulty TRX32 card. This does not necessarily mean that the card needs to be replaced. Malfunctioning elements can be disabled. This will affect the instrument performance and the calibration results, but the magnitude of the affect depends on the positions of the malfunctioning or disabled elements in the matrix. Elements near the centre of the array have a larger affect.
4.2.3.1.2 Noise test

The ME70 software includes a utility to measure noise. It tests that the electronic noise on the receiving boards conforms to the specification and helps to ensure that the system is running properly. Run the “Passive Test/Noise Level Determination” before calibrating the system.

1) Set “Range” to 300 m.
2) Set “Tx Mode” to Passive.
3) Position the cursor over the echogram for the most vertical beam and click the left mouse button to make this view active.
4) In the menu, open the “Extras” tab and read the noise value in dB ref 1W.

When the vessel is stopped and all other acoustic equipment is inactive, the noise estimate should be ≤ –145 dB re 1W.

4.2.3.2 Sphere

The ME70 should be calibrated using a 25-mm-diameter sphere made from tungsten carbide with 6% cobalt. A weight should be suspended at least two pulse lengths be-
neath the sphere to stabilize its position within the beam. Note that the apparent separation between the sphere and the weight decreases for measurements made in the most steered beams.

If fish or current are complicating the calibration, the calibration may be accomplished using both a 75.0- and an 84-mm diameter WC sphere. However, in that case, it is necessary to merge the results from the two calibrations. A Matlab script (concatenate_cal.m) is available that combines the measurements made with two spheres into one calibration data file.

4.2.3.3 Environment

The ME70 software has inputs for the water sound speed, \(c_w\) (m s\(^{-1}\)), at the transducer and throughout the sample range. The \(c_w\) at the transducer is used for beamforming. Measure the harmonic mean \(c_w\) for the sample range and input it to the ME70 software for estimating the range to each sample. If hydrographic conditions change significantly during the calibration procedure (e.g. indicated by a large tidal amplitude), the \(c_w\) measurements should be repeated during the calibration, and a revised harmonic mean \(c_w\) should again be input to the ME70 software.

The theoretical sphere target strength, \(TS_{\text{theory}}\) (dB re 1 m\(^2\)) depends on \(c_w\) at the depth of the sphere. The \(TS_{\text{theory}}\), specified in the ME70 calibration software, has been computed by Simrad for \(c_w = 1490\) m s\(^{-1}\). Generally, \(c_w\) will be a different value, and the calibration parameters must be adjusted accordingly. A Matlab script (compute_ME70_TS_Sphere_calibration.m) is available to generate \(TS_{\text{theory}}\) files for different values of \(c_w\).

4.2.3.4 Sphere movement

When calibrating each beam, the sphere should always be in the transducer far-field. For vertical beams, the sphere range will equal the sphere depth minus the transducer depth. For steered beams, the sphere depth may be reduced while maintaining the far-field requirement.

Depending on the ease of obtaining sphere-echo detections in the most steered beams, either:

- Calibrate one of the most steered beams and then calibrate the others, moving sequentially from one side of the swath to the other; or
- Calibrate the most vertical beam and then calibrate the others on one side of the swath, moving sequentially to the most steered beam. Return the sphere to the most vertical beam and then calibrate the beams on the other side of the swath, moving sequentially to the most steered beam.

In either case, if the sphere range changes, adjust the minimum and maximum depth of the detection window.

4.2.3.5 Software

The ME70 software includes a calibration utility.

1) In the “C:\Program Files\Simrad\SMS\ME70\CalibrationReports” directory, create a subdirectory (e.g., named YYYYMMDD_SURVEY_CAL_SITE) where the report will be saved.

2) In the “Setup\Install\Environment” tabs, input the water sound speed at the target depth in the “Water Column” tab; and input the water sound speed at the transducer face in the “Transducer Face” tab. Alternatively, input a sound speed profile.
3) In the “Operation\View Beam mode\Fan of Beams” tab, check that all of the beams are split beams.

4) In the “Operation\View Beam mode\Calibration” tab, check that calibration has been set to “Nominal Performance”.

5) In the “Setup\Data output” menu, select “Record File”. In the proposed data record directory, create a subdirectory with a descriptive name for the calibration raw data. These data may be used to later replay the calibration. Set an explicit “File Name Prefix” in the “Raw Data” tab, including, for example, the name of the configuration to be calibrated. Set the acquisition “Range”.

6) Move the sphere in the central beam at the required depth and note the minimum and maximum depth of the desired detection window. This will be used later to set the “File/Setup” window parameters.

7) Start the calibration utility by selecting “Calibration” in the “Setup” menu.

8) In the “File/Setup” menu of the “Calibration” Display, set the following parameters:
   - “Name” selects the calibration sphere used.
   - “Max. Deviation” defines, for TS data collection, the maximum allowable difference between the measured and the theoretical TS. Default = “10 dB”.
   - “Target Depth Limits, Upper” sets the upper limit for TS data collection. All detections at smaller depths will be ignored.
   - “Target Depth Limits, Lower” sets the lower limit for TS data collection. All detections at larger depths will be ignored.
   - Click “OK”.

The “Single Target Detection” window is available during calibration. To minimize erroneous targets (e.g. the weight beneath the sphere), the “Upper and Lower Target Depth Limits” should be frequently adjusted according to the sphere depth.

9) In the “File/Set-TS Detection” menu of the calibration display, set the following parameters:
   - “Min. Threshold”, The minimum TS detected. Typically set to the minimum theoretical sphere TS minus 8 dB. Lowering this parameter too much will decrease the detection performance.
   - “Min. Echo Length”. The minimum normalized echo length (duration of the echo pulse measured –3 dB relative to the peak divided by the duration of the transmitted pulse) to detect a target. Typically set to 0.8.
   - “Max. Echo Length”. The maximum normalized echo length to detect a target. Typically set to 1.8.
   - “Max. phase deviation”. The maximum standard deviation of split-beam phase sample in the echo pulse to detect a target. Typically set to 8.
   - “Max. gain Comp”. The maximum one-way beam-pattern compensation to detect a target. This parameter limits the maximum off-axis angle to which targets are detected. Typically set to 3 dB to allow target detections within the –3 dB one-way beamwidths. If the signal-to-noise ratio is high, then set to 6 dB to allow detections within the –6 dB one-way beamwidths.
   - “Min. echo spacing”. The minimum number of samples between echo pulses to detect a target. Typically set to 1.
   - Click “OK”.

Concurrently display both the sounder and calibration windows to observe the swath during the entire calibration procedure.

10) In the calibration display, select the “Result” tab.

11) In the calibration menu, begin the calibration by clicking “Start”. Move the sphere to obtain approximately 100 measurements distributed throughout each beam, particularly in the athwartships plane.

To reduce variability in the sphere-\(TS\) measurements and post-processing time, stop the collection of calibration data during noisy periods (e.g. echoes from bubbles or animals).

13) In the display, locate and set the “Upper Depth” and “Lower Depth” ranges. Update these ranges when the sphere detections too closely approach these detection-range limits.

A Matlab script (clean_up_cal.m) is available to filter the sphere detections. To use this script, note the start and end times when the sphere is in each beam. The script will filter detections outside these periods and thereby save post-processing time.

14) When the data acquisition is complete, click the “Stop” button in the calibration menu and analyse the detection results for each beam.

In the “Result” tab, identify beam models with standard deviation values above 0.4 dB.

In the calibration “Fan View”, select the beam to be analyzed.

Select the “TS-Data” tab to observe information about the target detections including detection “Number, Time, Depth” (= transducer depth + range from transducer to sphere), “TS-Comp” (target strength), “TS-UnComp” (target strength plus twice the directivity pattern), “Along[ships]” and “Athwart[ships]” beam angles (target position in the beam), and “\(S_a\)” (integrated volume backscattering coefficient for the target). Locate the invalid data through the sorting tools available for each column. For example, sort the data by “TS-Comp” or “Depth” values. The red and blue lines indicate the highest and lowest “TS-Comp” values, respectively.

Filter the invalid detections with the “Suspend” tool. The red and blue lines are refreshed.

15) In the “File” menu of the calibration display, select “Save as” to save the calibration data.

In the “C:\Program Files\Simrad\SMS\ME70\CalibrationReports” directory, select the directory in which to save the data. Click “OK”.

Name the file with the current time. “Indicated name” will be automatically augmented to read “IndicatedName_ConfigurationName_Date.”

In the “Description” field, indicate the calibration area and seabed depth; sphere material, size, and identification; weather conditions; post-processing detail; and any other relevant information. Click “OK”.

16) If invalid detections have been removed (e.g. echoes from fish or the weight) answer “Yes” to “Do You want to save the calibration to the current beam configuration?” Otherwise, if more filtering is required, answer “No” and perhaps use the Matlab script (see Step 13) to filter invalid detections.

17) Answer “Yes” or “No” to the option for generating a calibration report (PDF format).
18) In the “File” menu, click “Exit” to quit the calibration utility.

19) In the “Operation/View Beam Configuration/Calibration” tab, check that the calibration gains have been updated.

### 4.2.3.6 Analysis

If there are additional invalid detections to be removed, post-process the calibration file. To do so, in the “File” menu of the calibration display, select “Replay XML” and select the recorded calibration report (Step 15). In the calibration display, click “Start” and follow the previously described procedure starting at Step 11. Pay attention to the target depth limits indicated in the “File/Setup” menu. It filters the target detections when clicking “Start”.

If the target detection settings must be refined (e.g. \( TS \) values or depth limits), replay the raw data files and reprocess the entire calibration starting at Step 7.

### 4.2.3.7 Documentation

Save the calibration report (.xml format) that is located in the “C:\ Program Files\ Simrad\ SMS\ ME70\ CalibrationReports” subdirectory. It allows filtering of the target detections off-line, if needed, and contains the following calibration information:

- sounder configuration parameters (e.g. number of beams, frequencies, powers, beamwidths, steering angles);
- sphere type(s);
- target detections for each beam, with their range, compensated and un-compensated TS, angles within the beam, and selected or suspended status; and
- calibration results (e.g. \( \Delta G_0 \) and \( \Delta S_{a\ corr} \) for each beam and measurement standard deviations).

Save the sample power and split-beam angle data (.raw format) acquired during calibration. These files may be replayed, e.g. with different target detection parameters, to redo the analysis, if necessary.

In addition to the acoustic data collected during calibration and the resulting values for \( G_0 \) and \( S_{a\ corr} \) archive the following information:

- \( c_w \) values or profile(s);
- area name and geographic coordinates;
- weather conditions;
- configuration filename;
- sphere identification(s);
- \( TS_{theory} \) for each beam frequency;
- BITE test screenshots, particularly the Matrix view during transmission phase; and
- indices of disabled transducer elements.

Summarize the calibration results in spreadsheet (e.g. Table 4.9) to archive, compare with the results from previous calibrations, and evaluate any changes in the system performance. For example, plot \( \Delta G_0 \) and \( \Delta S_{a\ corr} \) vs. beam steering angle for successive calibrations of the same configuration.

**Table 4.9. Example of calibration compilation results.**

<table>
<thead>
<tr>
<th>Date</th>
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<tbody>
<tr>
<td>Vessel</td>
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</tbody>
</table>
### Experimental Setup

| Configuration name | Calibration area | Weather condition | Sphere type | $c_w$ at transducer | $c_w$ at sphere | Disabled elements | Comments: | Beam number | $f$ (Hz) | Steering Angle | $TS_{theory}$ | No. of detections | $TS$ std. dev. | $G_0\text{nom}$ (dB re 1) | $\Delta G_0$ (dB re 1) | $G_0\text{nom} + \Delta G_0$ | $\Delta S_{a\text{corr}}$ (dB re 1) | Previous $\Delta G_0$ |
|--------------------|------------------|-------------------|-------------|---------------------|----------------|-------------------|------------|-------------|--------|------------------|----------------|----------------|----------------|----------------|-----------------|----------------------|----------------------|------------------|-----------------------------|
|                    |                  |                   |             |                     |                |                   |            | 1           |        |                 |                |                |                 |                |                 |                    |                      |                 |                |
|                    |                  |                   |             |                     |                |                   |            | 2           |        |                 |                |                |                 |                |                 |                    |                      |                 |                |
|                    |                  |                   |             |                     |                |                   |            | 3           |        |                 |                |                |                 |                |                 |                    |                      |                 |                |
|                    |                  |                   |             |                     |                |                   |            | 4           |        |                 |                |                |                 |                |                 |                    |                      |                 |                |
|                    |                  |                   |             |                     |                |                   |            | 5           |        |                 |                |                |                 |                |                 |                    |                      |                 |                |
|                    |                  |                   |             |                     |                |                   |            | 6           |        |                 |                |                |                 |                |                 |                    |                      |                 |                |
|                    |                  |                   |             |                     |                |                   |            | 7           |        |                 |                |                |                 |                |                 |                    |                      |                 |                |
|                    |                  |                   |             |                     |                |                   |            | 8           |        |                 |                |                |                 |                |                 |                    |                      |                 |                |
|                    |                  |                   |             |                     |                |                   |            | 9           |        |                 |                |                |                 |                |                 |                    |                      |                 |                |
5 Emerging protocols

5.1 Multibeam echosounders

A multibeam echosounder that is designed principally to collect bathymetry data is often called a “hydrographic” (Lurton, 2010) multibeam echosounder. Hydrographic multibeam echosounders inherently measure backscatter intensity while generating depth measurements, and most modern systems allow some or all of these echo intensities to be recorded. These backscatter intensities can coincide with the seabed or with objects in the water column. They can be used to estimate $s_v$, $\sigma_{bs}$, $s_v$, or $s_a$ of fish, provided that some level of intensity calibration is ensured. The capability of hydrographic multibeam for recording backscatter intensity has improved as users increase demand for these data for a variety of purposes (see Brown and Blondel, 2009; Colbo et al., 2014).

Meanwhile, fishery multibeam echosounders have emerged that are designed principally to collect backscatter intensity data, but can also collect bathymetry data (see Section 4.2). Currently, both hydrographic and fishery multibeam echosounders are used to investigate fish and their seabed environments.

5.1.1 Requirements

The utility of data from either class of multibeam echosounder is increased if the instrument is calibrated to some level.

5.1.1.1 Level 1

Foremost, a multibeam echosounder must clearly image the surveyed seabed, with account for specular reflection at normal incidence and decreasing intensity with increasing grazing angle. The echo-intensity data should be compensated for the effects of transmit power and pulse duration, beam-directivity patterns, and gains. The first level of calibration uses nominal values based on the system design and specifications.

5.1.1.2 Level 2

A multibeam echosounder may also be used to measure variations in seabed reflectivities on various spatial scales. The second level of calibration must ensure that the system performance is stable vs. time (i.e. within and between survey periods) and the survey environment, e.g. water temperature and depth (Beamiss et al., 2002). The stability of the multibeam echosounder performance must be systematically and periodically evaluated relative to the same reflector (e.g. a reference area of seabed; see Section 5.1.4).

5.1.1.3 Level 3

The highest level of system calibration ensures that the stability and accuracy of the backscatter intensity measurements is sufficient to compare data collected with other calibrated echosounders and to invert volume- or seabed-backscatter models to estimate physical properties. In this case, the performance stability of the multibeam echosounder must be systematically and periodically evaluated relative to a target chosen for its well characterized and stable backscatter, e.g. a sphere (see Chapter 2).

5.1.2 Measurements

Crude measurements of bathymetry (i.e. range from the transducer to the seabed) only require an estimate of the mean sound speed of water between the transducer and the seabed. This value is used to convert the two-way sound-propagation time to an estimate of the range to the seabed from the transducer. Measurements of bathymetry for
hydrography account for (i) the geographic position and time from a GPS receiver, (ii) sound speed vs. depth from a sound speed or conductivity-temperature-depth (CTD) probe, (iii) sound refraction (important if not projecting vertically), (iv) transducer movement (e.g. roll, pitch, yaw, and heave) from an attitude sensor, and (v) installation geometry of the transducer and ancillary sensors. The offset distances between the transducer and ancillary sensors, and the roll, pitch, and yaw biases, are estimated using a “patch test.” This involves transects over flat areas and areas with a known feature, such as a slope or object on the seabed (e.g. pipeline or solid block). The accuracy of the bathymetric measurements is estimated from a model that combines the systematic error estimated for each measurement component.

Many multibeam echosounders also record backscatter intensity. Although measurements of bathymetry and uncalibrated measures of backscatter intensity are often used to classify seabed, those classifications are only comparable to measurements made at similar ranges with the same system and settings (e.g. beamwidth, gain, and transmit frequency, power, and pulse duration) (ICES, 2007). Therefore, it is important to calibrate multibeam echosounders to a high level if they will be used to measure surface backscattering coefficient, $s_s \left( \text{m}^2 \text{m}^{-2} \right)$, or volume backscattering coefficient, $s_v \left( \text{m}^2 \text{m}^{-3} \right)$, or both (Cutter et al., 2010), enabling comparisons of measurements with empirical or theoretical models for backscatter from fish (Foote, 1991b) or the seabed (Jackson and Richardson, 2007).

A high-level instrument calibration confirms that the multibeam echosounder is functioning properly and the data are accurate. Accurate seabed backscatter data may be used to estimate seabed slope, roughness, rock or sediment type, grain size, density, and porosity (Cutter and Demer, 2014) and improve statistical seabed classifications (ICES, 2007) that are comparable among sonars and over time. However, instrument calibration does not validate the seabed classifications.

### 5.1.3 System performance

With the receiver and transducer decoupled, reference voltages may be input to each receiver channel to check its response and sensitivity. Electrical impedance or admittance of each transducer element in both the transmit and receive arrays can also be measured vs. frequency. These tests may also measure the electrical impedance or admittance of transducer elements vs. frequency. The test results may identify failing receivers or transducer elements, and provide measures of gain and dynamic range. Some multibeam echosounders include utilities for these tests. Although this approach may be expanded to test all of the components involved in the transmit-and-receive process, it is more prone to cumulative error than reference-target methods for measuring the system response. Nevertheless, such tests should serve to monitor the performances of system components between periodic system calibrations with a standard target.

### 5.1.4 Seabed reference

Selected regions of the seabed have long been used as level-2 (see Section 5.1) calibration reference targets to check the general operation of single-beam echosounders, assess the effects of the environment (Dalen and Lovik, 1981), and compare their performances between ships (Hjellvik et al., 2008). Ideally, a reference seabed would have a constant, omni-directional backscatter. To approximate these requirements, the seabed reference should be comprised of objects that (i) scatter, vs. reflect, sound; (ii) have high volume attenuation and low bioturbation; and (iii) are virtually void of infauna and
epifauna. These characteristics will minimize fluctuations in the backscatter (ICES, 2007).

The expected seabed backscatter may be estimated from geophysical parameters (e.g. grain size, roughness) and a theoretical scattering model or by measurements made with an independent calibrated system. Assuming omni-directional backscattering from a flat, homogenous region of seabed, Hellequin et al. (2003) estimated the directivity pattern and gain of a hydrographic multibeam.

### 5.1.5 Sphere calibration

Multibeam echosounders typically measure split-beam angle data in one plane (e.g. athwartships) to detect the range to the seabed, but these data may not be recorded. Therefore, to use the sphere method (Foote et al., 1987) commonly used to calibrate split-beam echosounders, the position of the sphere within the beam must be measured accurately and independently. Although this constraint is the same as for single-beam echosounders, multibeam echosounders have narrow beamwidths (e.g. 0.5–2°) that require the sphere position to be adjusted and known with higher accuracy (i.e. ≤0.1°). Despite these challenges, several protocols are emerging for calibrating multibeam echosounders.

The sphere method (see Chapter 2) has been used by Melvin et al. (2003) and Foote et al. (2005) to calibrate high-frequency multibeam echosounders in a sea well or test tank. They used a spherical target because its backscatter is the same for all acoustic incidence angles and, therefore, beam-steering angles. While adjusting the sphere depth, the multibeam transducer was accurately rotated to estimate the two-way beam directivities, alongships and athwartships beam angles, and the system gains (Figure 5.1).

![Figure 5.1. Schematic diagram of the Woods Hole Oceanographic Institution sea well, as configured for calibrating a Simrad SM20. The receiving elements are in the horizontal plane, with θ = h = 0. The rotational axis of the transducer is in the vertical plane.](image)

### 5.1.5.1 Receiver linearity

The linearity of a multibeam echosounder receiver may be evaluated by changing the system gain while observing the sphere echo at different ranges (Foote et al., 2005).
Alternatively, Greenaway and Weber (2010) used a seabed reference (see Section 5.1.4) and varied the source level and receiver gain settings throughout each of their respective ranges. The pulse duration could have also been modulated. Their results included the maximum linear receiver output at each gain in relative, system-specific units, and the relationship between the source level and gain settings and their actual values.

5.1.5.2 Transducer near-field

Generally, multibeam echosounders should be calibrated with the sphere in the transducer far-field. However, the near-field for a multibeam transducer may range from 10 m for high-frequency systems to several 10s–100s of metres (e.g. 500 m for an 8-m array at 12 kHz) for low-frequency systems. Also, some multibeam echosounders use dynamic focusing, i.e. beamforming of the received signals, to operate more effectively in the near-field. Near-field focusing can make the equivalent beam angles and the system gain dependent on range. Therefore, it may be necessary to calibrate some multibeam echosounders with the sphere in the transducer near-field.

In cases when the calibration can only be calibrated in the near-field or when the instrument is used to make measurements in the near-field, or both, the near-field response should be measured. This is done by positioning the sphere on the acoustic axis and measuring its echo at discrete positions. The beamformed results are plotted against the transducer target range. The measurements are used for comparison against computations based on the transducer geometry and transmit frequency. Account should be made for any beam focusing. Confirmation of the computational results provides support for the model, which can then be used to extrapolate to different ranges.

When calibrating a bistatic multibeam, it may be necessary or advantageous to orient the arrays so that their acoustic axes intersect at the target range. This is particularly important for reducing the effects of parallax in a near-field calibration. An easy method for adjusting the relative array orientations is to place shims between the bolts and mounting plate at the far end of one of the transducer arrays.

5.1.5.3 Directivity pattern

To measure the directivity pattern, suspend the sphere at a fixed distance from the transducer at the nominal depth of the transducer. Rotate the transducer to make measurements of the sphere until the echo amplitude steadily decreases and then rises, indicating that the measurements spanned the beam to the first null. Adjust the sphere depth by a distance corresponding to ~0.1-fold the nominal transducer beamwidth and repeat the measurements. Do this until the beam is mapped between nulls in the horizontal and vertical planes. Repeat for each beam.

From the data, for each beam, locate the beam axis and measure the beamwidth. Locate the centre of the swath by plotting the peak amplitudes of all of the beams. If needed, fit a quadratic or other nonlinear function to locate the centre. Repeat the measurement near this computed centre to precisely determine the acoustic centre and estimate any offset angles. If the acoustic centre is offset significantly from the geometric centre, the measurements should be repeated to ensure their precision.

5.1.5.4 Gain

With the sphere on the beam axis, measure the echo amplitude and compute the system gain (see Chapter 2). Repeat these measurements for each beam and gain and pulse duration setting. Periodic calibrations allow the system stability to be monitored. Note, however, that calibrations in a tank do not account for any effects of the transducer
platform used to conduct a survey (e.g. baffling by the hull, gondola or fairing, and diffraction from nearby objects (Lanzoni and Weber, 2010).

To apply the sphere method to hull-mounted multibeam echosounders, a more elaborate system is needed to accurately move the sphere under the vessel at large ranges. For example, Lanzoni and Weber (2011) explored the use of a split-beam echosounder mounted adjacent to a multibeam echosounder transducer to accurately position the target within the multibeam echosounder beams. In this case, the split-beam transducer must be (i) mounted rigidly and near the multibeam echosounder without affecting the multibeam echosounder's performance and (ii) moveable such that it can be used to position the sphere with each of the multibeam echosounder beams.

### 5.1.6 Simrad SM20

The Simrad SM20 (formerly the SM2000) is a multibeam imaging sonar that is available in two operating frequencies, 90 and 200 kHz. The system consists of a curved transmit-and-receive array (Figure 5.2). With an optional external linear transmit array, the two arrays form a Mills Cross. The radius of the curved array is 38.4 cm for the 90-kHz system and 20 cm for the 200-kHz system. It is comprised of 80 evenly spaced rectangular elements that create 128 beams spanning 94.2 and 88.2° for the 90 and 200 kHz systems, respectively. The external transmitter is 87 cm for the 90 kHz system and 42 cm for the 200-kHz system. When using the external transmitter, both systems have nominal azimuthal-beam resolutions of 1.5°.

![Figure 5.2. Photos of the SM20/200-kHz system. (A) Image mode (monostatic configuration). (B) Echosounder mode (bistatic configuration). (C) Sonar heads and desktop computer.](image)

Below is a summary of the results from sphere calibrations of two SM20 systems. A 90-kHz system operating in profiling mode (bistatic) was calibrated in both the sea well on the Woods Hole Oceanographic Institution (WHOI) Iselin Dock and in a freshwater tank (Chu et al., 2003). The targets were (i) 20- and 38.1-mm diameter spheres made from tungsten carbide with 6% cobalt, (ii) 23- and 60-mm diameter spheres made from copper, and (iii) a 60-mm diameter sphere made from aluminium. A 200-kHz system operating in imaging mode (monostatic) and profiling (bistatic) was calibrated by IRD
and Ifremer using a 20-mm diameter sphere made from tungsten carbide with 6% cobalt.

In the sea well, the 90-kHz SM20 system was mounted on the end of a shaft, aimed horizontally, and connected to a rotator. The rotator accuracy was ±0.1° and was controlled remotely either manually or using computer software (Matlab GUI script) (Figure 5.1). The calibration sphere, tied in a monofilament harness, was suspended from three monofilament lines controlled by motorized fishing reels in the far-field of the SM20 transducer (i.e. 23 m) at an initial depth of 3 m. During the calibration, the sphere remained in one position, and the transducer was rotated allowing all of the beams to scan the sphere. The sphere depth was then adjusted for successive scans to cover the entire width and length of the array.

The system directivity patterns are shown in the vertical plane at 23 and 11.7 m, respectively (Figure 5.3a and b), and in the horizontal plane (azimuth) at 11.7 m (Figure 5.4). The 1-D directivity patterns of the individual beams are shown in the equatorial plane (horizontal) in Figure 5.5 and its 2-D directivity pattern (central region) is shown in Figure 5.6.
Figure 5.3. Directivity pattern of a 90-kHz SM20 multibeam transducer in the vertical plane, as determined by changing the depth of a rubber-walled focusing sphere (a) at 23-m range in 4-cm increments in a sea well; and (b) at 11.7-m range in 2-cm increments in a freshwater tank. The maximum intensity value is that observed in a spatial window consisting of five contiguous beams and five range cells centred on the respective beam and range of nominal greatest sensitivity. The mean is that of all echo values within the window. A Gaussian function is fitted to the maximum intensity values.
Figure 5.4. Measured equatorial-plane directivity pattern of a 90-kHz SM20 multibeam echo-sounder. The target range was 11.7 m.

Figure 5.5. Equatorial-plane directivity patterns, measured and estimated, for a set of individual elements of a 90-kHz SM20 multibeam transducer. Measured values of the beamwidth (BW) are shown.
Figure 5.6. Measured 2-D directivity pattern in the central region of the beam for a 90-kHz SM20 multibeam transducer at: a) ~23-m range in the sea well; and b) ~11.7-m range in a freshwater tank. Echo levels spanning 21 to 0 dB are dark red.

The directivity patterns, horizontal (azimuth) and vertical (elevation) planes of the 200-kHz SM20 operating in imaging mode (monostatic), are shown in Figure 5.7.

Figure 5.7. 200-kHz SM20 transmit directivity pattern and sensitivity (dB ref 1 µPa/V at 1m), measured with the hydrophone (pulse duration: 250 µs, power: high) in Imaging (left) and Echo-sounder modes (right).

5.2 Wide-bandwidth echosounders

Narrow-bandwidth echosounders transmit and receive finite-duration pulses of a continuous sinusoidal wave (CW) (Figure 5.8). A Fourier transform of the transmitted or
received signal indicates its frequency spectrum. Due to the pulse duration and system filters, the spectrum of a CW pulse is not entirely at the centre or carrier frequency, but is spread among the surrounding frequencies, the range of which is characterized as bandwidth (e.g. Section 2.4.4 in Simmonds and MacLennan, 2005).

The quality factor, Q, is defined as the ratio of the half-power bandwidth to the centre frequency. A higher Q indicates a narrower bandwidth and a lower Q indicates a broader or wider bandwidth (Figure 5.8). Referring to bandwidth, wide and broad are synonymous. Narrow and wide (or broad) refer to bandwidths that are a few percent and ≥10% of the centre frequency, respectively.

Wide-bandwidth echosounders transmit and receive finite-duration pulses of a widebandwidth signal. A linear frequency modulated chirp signal is commonly chosen as a transmit signal because it is simple and provides adequate resolution and bandwidth. The wider the bandwidth, the more information is included in the backscatter spectra to potentially improve target classifications (e.g. Holliday, 1972b; Simmonds, 1990; Simmonds et al., 1996; Zakharia et al., 1996). A wide bandwidth also serves to improve estimations of fish sizes, speeds, and orientations (Holliday, 1974; Nero and Huster, 1996; Nero et al., 1998; Stanton et al., 2010). Despite their potential benefits, wide-bandwidth echosounders have not been used routinely in fishery surveys because they have not delivered sufficient power over the bandwidth to sample the requisite observation ranges. Explosives and arcers can produce high source levels with wide bandwidths, but they are dangerous and their signals are variable and unpredictable (Holliday, 1972b). Also, the receive sensitivity per frequency band is less in wide-bandwidth transducers compared to narrow-bandwidth transducers. Additionally, wide-bandwidth-system design is complicated. Nevertheless, as electronics get smaller, more powerful, and more reliable, commercial wide-bandwidth systems are being developed (Jones et al., 2009). Consequently, the use of wide-bandwidth acoustic instrumentation is again receiving attention (Conti and Demer, 2003; Conti et al., 2005b; Au and Benoit-Bird, 2008; Benoit-Bird et al., 2008; Stanton et al., 2012; Forland et al., 2014).

Wide-bandwidth echosounders can be calibrated using the methods described in Chapters 2 and 4. As with narrow-band echosounders, the sphere method is recommended. Additional considerations for calibrations of wide-bandwidth echosounders
include the frequency dependence of many variables in the sonar (or radar) equations (see Chapter 1) and the sphere TS.

The received power, $p_{er}$ (W), wavelength, $\lambda$ (m), absorption coefficient, $\alpha_a$ (dB m$^{-1}$ re 1), and equivalent two-way beam angle, $\psi$ (sr), are all dependent on frequency. Additionally, due to the transmit signal and transducer response, the source level varies with frequency (Figure 5.9). The elements within a transducer have a natural resonance frequency, and a bandwidth over which electrical and acoustic energy can be efficiently converted. The quality factor $Q$ of the transducer describes the shape of the frequency pass-band. Low $Q$ transducers have wide bandwidths. Optimally, the transmitting and receiving responses are “flat” over the bandwidth such that the spectral shape of the received signal is only a function of the response of the target. Generally, however, the transducer transmitting and receiving responses are not flat and must be measured empirically and taken into account. The transducer transmitting and receiving response functions are usually measured once in controlled, high signal-to-noise ratio (SNR) conditions (i.e., in a tank) before the transducer is mounted and then not measured again, at least not routinely, throughout the life of the transducer. For calibration and processing purposes, these response functions are input to the algorithms and used to derive the receive spectrum. Similar to narrow-band systems, the combined transmit and receive response function is often measured and used rather than using separate transmitting and receiving functions. The quality of the response functions is monitored over time by evaluating the received spectra of $p_{er}$ from a calibration sphere. Changes in the spectral shape of $p_{er}$ that cannot be explained by changes in the sphere, environmental factors, or other system components may be due to degradation of the transducer. In that case, the transducer should be independently evaluated.

The frequency dependence of the directivity pattern has substantial effects on off-axis measurements of $\sigma_{bs}$ measurements and $\psi$ (Lavery et al., 2010a, 2010b). For a wide bandwidth transducer having a constant active area, its beamwidth is inversely proportional to frequency. For example, if the highest frequency equals twice the lowest frequency, then the beamwidth for the highest frequency will be half that for the lowest frequency.

The frequency-dependent beam patterns of split-beam transducers can be well approximated using split-beam estimates of off-axis angles, $\alpha, \beta$ (°). Using the sphere method (Chapters 2 and 4), the combined transmit-receive directivity pattern, $b(\alpha, \beta)^2$ (dimensionless), beam widths, $\alpha_{-3\ dB}$ and $\beta_{-3\ dB}$ (°), and axis offsets, $\alpha_o$ and $\beta_o$ (°) can be estimated. To make these measurements of a single-beam transducer, it is necessary to have independent motion control of the transducer and sphere. It is possible to provide independent measures of sphere location from a concurrently operated multibeam, split beam, or lens system.

A major challenge to sphere calibrations of wide-bandwidth echosounders involves the consistency and prediction of $\sigma_{bs}(f)$ over the bandwidth of the system (Stanton and Chu, 2008). The $\sigma_{bs}(f)$ for even simple targets such as spheres have numerous resonances that are sensitive to the material properties of the target as well as the environmental conditions (see Chapters 2 and 3). These resonances are caused by the interference among the echoes from the front interface, internally refracted and reflected waves, and circumferential waves (Hickling, 1962; MacLennan, 1981).

For narrowband systems, resonances in $\sigma_{bs}(f)$ can be avoided by judicious choices of material and sizes of spheres. This may not be possible for wide-bandwidth systems. Although the sharp nulls provide well-defined features to compare the theoretical and measured spectra, their positions are sensitive to material properties of the sphere and
sound speed and density of the water. Therefore, the null positions may not be sufficiently well known to use for reliably calibrating $\sigma_{bs}(f)$ (Lavery et al., 2010b; Hobaek and Forland, 2013). In addition to robust features in the sphere scattering spectra, it is advantageous to have flat or rising portions of the spectral response curve at the band edges. Such characteristics at the edges of the theoretical sphere backscattering spectra serve to better monitor the edges of the system bandwidth.

### 5.3 Partial-wave calibration

The sphere calibration described in Chapters 2 and 4 uses theoretical and measured values of sphere $\sigma_{bs}$ derived from the entire backscattered wave. To calibrate widebandwidth echosounders, Stanton and Chu (2008) introduced a sphere calibration technique that uses only part of the wave backscattered from the sphere. This partial-wave $\sigma_{bs}$ takes advantage of the fact that the echo from the front interface is relatively insensitive to frequency (Dragonette et al., 1981) and that this component can be separated from the total echo using pulse-compression techniques (Figure 5.9) (Chu and Stanton, 1998). In practice, this calibration technique is robust and reliable, even when spheres on the order of 10s of cm in diameter are used to calibrate systems in the 1-10 kHz range (Stanton et al., 2010).

![Figure 5.9. Effects of pulse-compression processing of a wide-bandwidth signal. (a) A 2-ms chirp transmit signal (raw); (b) its frequency spectrum; (c) the envelope of its pulse-compressed signal, achieved by autocorrelation or matched filtering (MF); and (d) the frequency spectrum of its pulse-compressed signal. The blue dots indicate the frequency spectrum calculated using a 300-μs-long window.](image)

The range resolution of a narrow-band echosounder is dependent on the pulse duration, and targets must be separated by greater than half the pulse length ($c_\omega \tau/2$) to be resolved. For wide-bandwidth systems, range resolution is inversely proportional to bandwidth (Ehrenberg and Torkelson, 2000). With increased bandwidth, finer range
resolution is achieved through pulse compression techniques (Chu and Stanton, 1998), a form of matched filtering. For example, as a result of pulse-compression processing, a 2-ms chirp, which corresponds to a narrow-bandwidth range resolution of 1.5 m, can be compressed to a signal with a duration of about 11.4 μs (Figure 5.9) and a range resolution of about 8.6 mm. This fine range resolution can facilitate a partial-wave calibration (Stanton and Chu, 2008; Stanton et al., 2010) by resolving the echoes from the front of a sphere and the latent circumferential waves. Thus, pulse-compression techniques can be used to differentiate full wave and partial wave contributions to the spectrum of backscattering as shown in Figure 5.10.

Figure 5.10. Acoustic backscatter from a 38.1-mm diameter tungsten carbide sphere. (a) The full-wave sphere-scattering amplitude (thin curve) and the partial-wave sphere-scattering amplitude for the echo from the front interface of the sphere (bold curve), each normalized by half the sphere radius. The circles indicate values at 38 and 120 kHz. (b) The impulse response of backscattering from the sphere. The impulse response of the echo from the front interface was used to define the partial wave scattering amplitude.

5.3.1.1 Example partial-wave calibration

A wide-bandwidth echosounder system housed in a towed body was calibrated in 2006 (Stanton et al., 2010). During the calibration, the system, normally towed in surveys with its acoustic beams aimed downward, was suspended 10 m directly below the vessel, which was anchored. The system was tethered so that the acoustic beams were aimed in the downward direction. Several calibration spheres were used, one at a time, suspended 30 m below the towed body with a weight positioned ~10 m below the sphere. The wide-bandwidth systems did not have split-beam transducers, so single-beam calibration methods were used.

The towed body, sphere, and weight were all free to move, which fortuitously moved the sphere in the beam. Account was taken for the pitch and roll of the tow body, which caused significant variability in the echo level. The acoustic data were indexed with values of heave, pitch, and roll from a motion sensor. Echoes with pitch and roll values near 0° were assumed to be “on-axis” and were used for analysis.
Figure 5.11. Time-domain analysis of an echo from a 41.5-cm-diameter air-filled aluminum sphere suspended 30 m below the towed-body. (Top) Envelope of the pulse-compressed echo. The echo at 0 ms is from the front interface and the smaller echo at ~0.6 ms is an echo from a circumferential wave (Lamb wave). (Middle) Fourier transform (impulse response) of the exact modal series solution for the sphere echo. (Bottom) Envelope of the impulse response convolved with the transmit signal. Each plot is normalized to unity.

The transmit signal used in the pulse-compression processing was measured from the reflection off an air–water interface. The compressed pulse resolved various features in the backscatter from the sphere (Figure 5.11). The partial-wave and full-wave backscatter signals were extracted by gating the signal in time such that the echo from the front interface (300-μs gate) and the entire sphere (3-ms gate) were resolved both temporally and spectrally. Each period was zero padded to form 4-ms time-series. For each series, the backscatter signals were obtained using an inverse Fourier transform over the appropriate bandwidth. An amplitude-weighting function was used to suppress side lobes.

The results from the two time-gates were compared (Figure 5.12a). The spectrum of the echo from the front interface is smoothly varying over the entire bandwidth. The spectrum of the entire echo (i.e., front interface and circumferential waves), is rapidly oscillating over the bandwidth. This observation is consistent with the theoretical predictions shown in Figure 5.9.

The system response of this wide-bandwidth echosounder was then determined from an analysis of the partial-wave backscatter spectra (Figure 5.12). This example shows that the partial-wave technique is less sensitive to uncertainties in sphere material properties and dimensions and may, therefore, significantly reduce the error in sphere calibration of a wide-bandwidth echosounder. Furthermore, due to the high
range resolution provided by pulse compression, the diameter of the calibration sphere may be minimized.

Figure 5.12. Spectra of signals and system response associated with a calibration performed with an air-filled spherical shell. (Upper) Spectra of the full-wave backscatter (thin line), and the partial-wave backscatter from the front interface (thick line). The spectra were calculated using the pulse-compressed signal illustrated in the bottom panel of Figure 5.11. (Lower) System response of the wide-bandwidth transducer as derived through the partial-wave analysis.

5.4 Acoustic Doppler current profilers

Since the mid-1980s, acoustic Doppler current profilers (ADCPs) have been fitted to many research vessels (Brierley et al., 1998b) and other platforms, and ADCP measurements have become an integral component of physical oceanographic studies (Woodward and Appell, 1986). Concurrently, ocean observatories have become key tools to monitor global climate change and its effects, as well as to provide indicators
of ecosystem status and trend. Therefore, it is worth considering how calibrated ADCP data may be used to complement echosounder data. Relative to data from scientific echosounders, ADCP data are generally limited by:

- larger vertical resolution;
- larger blind zones near the sea surface and seabed;
- lower receiver dynamic range;
- temperature sensitive system performance;
- multiple, non-vertical single beams; and
- difficult sphere calibrations (due to non-vertical single-beams).

Notwithstanding these limitations, ADCPs can provide measures of current velocity, and fish and zooplankton velocity (Demer et al., 2000; Zedel et al., 2003, 2005; Zedel and Cyr-Racine, 2009) and volume backscattering coefficient, \( s_v \) (m\(^2\) m\(^{-3}\)) (Brierley et al., 1998a). The calibration protocol in this section is relevant to the measurements of \( s_v \).

AdCP measurements of \( s_v \) may be validated using one of three methods:

1) Estimate the sonar-equation parameters (see Chapter 1, component calibration);
2) Compare the measurements with those from a calibrated echosounder (see Chapter 1, comparison method); or
3) Compare the measurements with biomass metrics from net samples or the output of backscatter models parameterized with animal species, sizes, and abundance estimated from net samples.

### 5.4.1 Component calibration

In commonly used ADCPs (e.g. Teledyne RD Instruments or RDIs), volume backscattering strength, \( S_v \) (dB re 1 m\(^2\) (4\(\pi\) m\(^3\))\(^{-1}\)), is calculated for data from each of four acoustic beams using the following sonar equation:

\[
S_v = 10 \log_{10} \left( \frac{4.47 \times 10^{-20} K_c K_t (t_w + 273) (10 K_r E_r - E_t)/10 - 1)^2}{c_w P K_t 10^{-4} a^{7/5}} \right),
\]

(5.1)

where \( t_w \) (°C) is the water temperature at the transducer, \( r \) (m) is the range along the beam to the scatterers, \( a_a \) (dB m\(^{-1}\)) is the absorption coefficient of water, \( P \) (m) is the transmit pulse length, \( c_w \) (m s\(^{-1}\)) is the mean sound speed between the transducer and the scatterers, \( K_t \) (W) is the beam specific transmitted power in the water (i.e. \( P_{at} \)), \( E \) (counts) is the received acoustic intensity, \( E_r \) (counts) is the reference acoustic intensity, \( K_c \) (dB count\(^{-1}\)) is the beam specific count conversion factor (aka RSSI), \( K_r \) (dimensionless) is the system noise constant, and \( K_t \) (dimensionless) is the frequency-specific system constant. Note that the units and terminology in Equation (5.1) are different from those in the other sections of this document. Also note that the expression for \( K_t \) in Zhou et al. (1994) and Heywood (1996) was corrected in Wade and Heywood (2001) as used in Fiedler et al. (1998), Plimpton et al. (2004), Lee et al. (2004), and Radenac et al. (2010). Some of the parameters in Equation (5.1) are nominal values from the manufacturer; the others must be measured. For mobile ADCPs, the parameters in Equation (5.1) may be measured in a test tank using a calibrated transducer (Tessier, 2006; Tessier et al., 2008).

To measure backscatter intensity using either a broad-bandwidth ADCP (BB-ADCP) or a narrow-bandwidth ADCP (NB-ADCP), Deines (1999) proposed a "working version" of Equation (5.1).
\[ S_v = C + 10 \log_{10} \left( \left( T_x + 273.16 \right) R^2 \right) - L_{DBM} - P_{DBW} + 2\alpha_a R + K_c (E - E_r), \]

where \( C \) accounts for the transducer efficiency, noise bandwidth, noise factor, transducer diameter, and transducer beam angle, \( L_{DBM} = 10 \log_{10}(p) \); and \( P_{DBW} = 10 \log_{10}(K_1) \). Note that (Gostiaux and van Haren, 2010) identified a typographical error in Equation A-5 in Deines (1999) and proposed a new formulation that may be less biased for measurements with low signal-to-noise ratios (snr). Table 1 in Deines (1999) provides values for \( C \) (accurate to ±3 dB), \( P_{DBW} \), and Rayleigh distance, \( R_0 \), for multiple RDI ADCPs. A BB-ADCP provides more precise measures, 1 dB vs. 5.6 dB for both current velocity and backscatter intensity, respectively, and their performances are less sensitive to temperature than to an NB-ADCP (Deines, 1999). Irrespective of the type of ADCP used, measurements should be made in the transducer far-field (Chapter 4). The \( \alpha_a \) values should be estimated from CTD profiles in the survey environment. The \( E_r \) value can be estimated as the counts measured at the largest recorded range, when noise dominates any remaining signal. If there is no effect of ship speed on \( E_r \), its value may be estimated while the vessel is stationary (Brierley et al., 1998b).

ADCP measurement of \( S_v \) may be more precise if the data span large spatial scales (Postel et al., 2007). Variation in measured \( S_v \) may be caused by the random positions and orientations of scatterers within the insonified volume, and noise. The \( S_v \) measured with each of the four ADCP beams may be averaged to improve measurement precision of randomly oriented organisms and to reduce measurement bias for polarized targets, e.g. schooling animals (Zedel et al., 2005).

### 5.4.2 Echosounder comparison

ADCP measurements of \( S_v \) may be compared to \( S_v \) measurements made with a calibrated echosounder and adjusted for observed differences. For example, Brierley et al. (1998a) recorded \( t_w \) throughout a survey and compared \( S_v \) measurements of krill made concurrently with a multifrequency (38-, 120-, and 200-kHz) scientific echosounder (Simrad EK500). They compared measurements from two krill layers, one shallow and homogeneously distributed, and the other deep and heterogeneously distributed (Figure 5.13).

**Figure 5.13.** Volume backscattering strength \( (S_v) \) measurements of krill made concurrently with a 153.6-kHz narrow-bandwidth ADCP and a calibrated multi-frequency (38-, 120-, and 200-kHz) scientific echosounder (Simrad EK500). The krill were aggregated (a) shallow and homogeneously distributed during night-time; and (b) deep and heterogeneously distributed during daytime. The x-axis labels are the same for both plots. The symbols indicate measurement at 120 kHz (solid line), ADCP (triangles), 200 kHz (squares), and 38 kHz (circles).
For the shallow layer, the measures from the two instruments were well correlated, except for $S_v$ at 200 kHz < \sim -80$ dB, where the echosounder measurement were dominated by noise. For the deep layer, the correlation was degraded, perhaps due to the patchiness of the scatterers, but mostly due to saturation of the ADCP receivers at $S_v > \sim -77$ dB. Similarly, Fiedler et al. (1998) compared measurement from a 150-kHz ADCP with those from a 38-kHz scientific echosounder (Simrad EK400) and found that the correlation decreased with increasing depth (i.e. $r^2 = 0.83$ at 25 m and 0.53 at 185 m), likely due to the decreasing snr. Lee et al. (2004) compared measurements of krill aggregations at various depths made concurrently with a 153.6-kHz BB-ADCP and a 200-kHz scientific echosounder (Simrad EK60) (Figure 5.14).

![Figure 5.14. Mean volume backscattering strength (MVBS) measurements of krill made concurrently with a 153.6-kHz broad-bandwidth ADCP (four-beam average) and a calibrated multi-frequency (200-kHz) scientific echosounder (Simrad EK60). The measurements were made while cruising (A) away from shore during day, and (B) towards shore during night. The lines indicate least-squares fits to the data.](image)

### 5.4.3 Net catch comparison

ADCP measurements of $S_v$ have been compared to (i) metrics of plankton biomass (e.g. displacement volume, dry weight, ash free dry mass) sampled in nets or (ii) $S_v$ estimated from models parameterized with data from net samples. For these comparisons, account must be made of the volumes sampled by the ADCP and the net.

Wade and Heywood (2001) compared $S_v$ measured with a 153-kHz ADCP (RDI) to catches in a 200-µm mesh net and estimates of dry-weight by size fractions. The correlations were highest ($r^2 = 0.76$) for the largest (500–2000 µm) size fraction (Figure 5.15). Lynn (2003) compared $S_v$ measured with a 150-kHz ADCP (RDI) to displacement volume (after removal of large gelatinous plankton) in samples from an oblique 0.505-mm-mesh bongo net (Figure 5.16).
Figure 5.15. $\log_e(\text{plankton volume})$ (cm$^3$ 1000m$^{-3}$) vs. mean volume backscattering strength (MVBS; dB) measured concurrently during two cruises (filled and open circles). The least-squares-fit line is for the combined data.

Figure 5.16. Theoretical mean volume-backscattering strength (MVBS; dB), calculated with a straight cylinder model with data on MOCNESS catches of euphausiids, and MVBS measured with an ADCP. Also shown are the ideal 1:1 relationship (dashed line) and the regression (solid line).

Postel et al. (2007) compared $S_v$ measured at short ranges with two 300-kHz BB-ADCPs to ash-free dry mass measure with 55- and 200-µm plankton nets. Generally, the correlations increased with size fraction (except the largest size class with a diverse species composition) and layer density, i.e. for the highest snr (e.g. Wade and Heywood, 2001).

Takikawa et al. (2008) compared $S_v$ measured with a 38-kHz BB-ADCP (RDI) dry-weight samples from a 0.33-mm-mesh net. Although nearly 90% of the dry weight was comprised of euphausiids, copepods, and appendicularians that are weak scatterers at 38 kHz, the correlation was highly significant.

Zhou et al. (1994) compared $S_v$ measurements made with a 153-kHz ADCP (RDI) to $S_v$ estimated from a straight-cylinder model parameterized using data from MOCNESS-net samples of euphausiids. They found large differences when the abundance is high (Figure 5.17), perhaps due to undersampling by the net.
Ressler (2002) compared $S_v$ measurements made with a 153-kHz NB-ADCP to $S_v$ estimated from scattering models for several species groups parameterized using data from MOCNESS-net samples of a variety of animals. The correlation was high for crustaceans, small fish, and fragments of non-gas-bearing siphonophores, but insignificant for gelatinous zooplankton, pteropods, atlantid molluscs, and gas-filled siphonophore floats.

Fielding et al. (2004), following (Flagg and Smith, 1989), compared $S_v$ measurements made with a 153-kHz NB-ADCP to $S_v$ estimated from scattering models for several species groups using data from 280-$\mu$m-mesh Longhurst-Hardy plankton recorder samples including amphipods, chaetognaths, copepods, euphausiids, fish, and pteropods. The correlations were better when only the most significant acoustic scatterers (amphipods, chaetognaths, copepods, euphausiids, fish, and pteropods) were considered. The species in the mixed assemblage did not contribute proportionately to the $S_v$, highlighting the need to account for species-specific backscatter when interpreting ADCP data (Figure 5.18).
Figure 5.18. Model-predicted backscatter mean volume backscattering strength (MVBS) vs. observed MVBS for (a) copepods, (b) euphausiids and decapods, (c) amphipods, (d) chaetognaths, (e) fish, and (f) pteropods.

$S_v$, measured with a single-frequency ADCP cannot be used for quantitative abundance estimates of mixed-species aggregations, but they may indicate the general zooplankton distribution (Flagg and Smith, 1989; Heywood et al., 1991; Fielding et al., 2004).

5.5 Imaging sonars

Acoustic cameras are high-frequency multibeam imaging sonars with linear transducer arrays with narrow beamwidths. During the past decade, advances in beamforming technology and miniaturization of hardware components have yielded several acoustic cameras (e.g. DIDSON, ARIS, BlueView, and Kongsberg M3) that may be used for fishery research. This section details some efforts to calibrate a Dual-frequency IDentification SONar (DIDSON).

5.5.1 DIDSON

DIDSON is a high-definition acoustic camera originally developed by the University of Washington’s Applied Physics Laboratory for application to military surveillance missions. The sonar and transducer array are well described by Belcher et al. (2001).

The DIDSON can collect multibeam data at two frequencies, depending on the desired target range and resolution. The high frequency (1.8 MHz) mode has high azimuthal (i.e. lateral, athwartships) resolution at small ranges (<15 m). The low frequency (1.1 MHz) mode has low azimuthal resolution at large ranges (15–40 m). The high frequency mode uses 96 beams, each $\phi = 0.3^\circ, \theta = 14^\circ$, where $\phi$ is the azimuthal beamwidth and $\theta$ is the elevation beamwidth. The total field-of-view is, therefore, $29 \times 14^\circ$, respectively. In low-frequency mode, each beam is $\phi = 0.4^\circ, \theta = 14^\circ$.

Depending on the mode and total range, data can be collected at 4–21 frames s$^{-1}$. Each frame is comprised of data from eight pulse sequences, each comprising 12 of the 96 beams. The data are stored as 8-bit (counts 0–255), $N_{\text{beam}} \times 512$-pixel images, where $N_{\text{beam}}$ is the number of beams. The total range equals the initial offset, the range from the active transducer elements to the first observation values, plus the observation “window” length. The range resolution equals the total range divided by 512 m pixel$^{-1}$. The value of $N_{\text{beam}}$ defines the swath width, the lateral extent of the data.
Rather than using digital beamforming, DIDSON uses acoustic lenses to form beams. The combination of a lens array and high operating frequency offer target resolutions that approach conventional optical methods (Simmonds and MacLennan, 2005). DIDSON images have been used for remote identification of targets based on morphology, behavior, and energetics (Moursund et al., 2003; Holmes et al., 2006; Mueller et al., 2006, 2010; Handegard et al., 2009). DIDSON also outputs estimates of target length and range, but these values are known to be biased (Burwen et al., 2010). Therefore, despite the many benefits of acoustic sensing in water and the high resolution two-dimensional (2-D) images it provides, DIDSON must be calibrated to provide more quantitative measures.

### 5.5.2 Backscatter calibration

A DIDSON can be calibrated for measurements of backscatter amplitude vs. range using methods similar to those used to calibrate single-beam echosounders (see Chapter 2). It is, however, very helpful to pair the DIDSON with a split-beam echosounder to monitor the sphere position. The DIDSON measurements of backscatter intensity, coupled with the split-beam measurements of target position, provide more accurate measures of backscattering cross section vs. off-axis angles. As described by Jech et al. (2012), the sphere positions identified by the split-beam were generally accurate to within 1–2 cm and precise to within a few millimetres.

Because the DIDSON beams are narrow, the sphere diameter must be much smaller than the beam cross section at the calibration range (see Chapter 4). This is so that the measurements of the sphere echo are made outside of the near-field range of the sphere, and the directivity pattern is approximately constant for the solid angle spanning the sphere. For example, if a 0.0381-m-diameter sphere was positioned in a 0.3° beam at 7.3 m range, the sphere would span the entire beam cross section. Moreover, the sphere echo would be sensed by multiple adjacent beams, causing spatial ambiguity.

The directivity patterns should be mapped for each lens configuration used. These maps can be used to correct image distortions caused by lens-specific imperfections (Negahdaripour, 2005). A nonlinear cubic distortion model may be applied to the horizontal field of view, but the model parameters are specific to each DIDSON and lens configuration (Negahdaripour, 2005; Negahdaripour et al., 2009).

### 5.5.3 Position calibration

To translate the positions of objects detected in 2-D DIDSON images to 3-D space, data is needed from two DIDSON having overlapping sample volumes. For example, (Negahdaripour et al., 2005) applied stereo-optical and machine-vision techniques to calibrate two DIDSONs in a stereo-pair configuration. Jech et al. (2012) mounted two DIDSON in a test tank such that the two arrays were rotated 90° relative to each other (Figure 7.19). They used two 38.1-mm-diameter spheres made from tungsten carbide with 6% cobalt to map the directivity patterns and a 120-kHz split-beam echosounder transducer (Simrad ES120-7) to provide independent estimates of the target positions.
Figure 5.19. (left) Two DIDSON mounted orthogonally, and a 120-kHz split-beam transducer (Simrad ES120-7). Within the volume sampled by both DIDSON, the units on the left and right provide information about the vertical (azimuth) and horizontal dimensions, respectively. (right) The array in a test tank at the University of New Hampshire.

The array was rotated, controlled by computer, while the spheres remained stationary. One sphere remained on the geometric centers of the DIDSON swaths, while the relative positions of the two spheres were measured by the DIDSON and split-beam systems. The 3-D positions of the spheres were estimated by merging the 2-D positions measured from the two DIDSON. The backscatter strength values were also recorded from each DIDSON.

Backscatter strength values (1–126; 93% between 50 and 105) were measured vs. location within the beams (Figure 5.20). For the horizontal and vertical DIDSON, the backscatter strength values measured near the edges of the beams were approximately 37 and 23% less than those measured near the beam axis, respectively. The overlapping field-of-view was approximately 14 x 14°, so the measured directivity patterns (Figure 5.20) do not represent the complete 14 x 28° swaths for each DIDSON.

Figure 5.20. Echo strength as a function of the sphere position measured by the horizontal (left) and vertical (right) DIDSON. The size of the circular symbol is proportional to the echo strength (values ranged from 1 to 126). Echo strength values, and their standard deviations in parentheses, are shown to the right of each measurement series.
5.6 Scanning sonar

Vertical echosounders sample beneath the ship, but do not detect marine life located close to the sea surface (Axenrot et al., 2004; Scalabrin et al., 2009; Totland et al., 2009). Scanning sonars, in comparison, sample a much larger volume, including the region near the sea surface. Scanning sonars may sample using multiple frequencies and provide a 360° radar-like scan from a moving ship. For example, the Simrad SH80 sonar operates at frequencies between 110 and 122 kHz (1 kHz step) and has 64 beams (Figure 5.21).

Despite these attributes, scanning sonars have rarely been used quantitatively (Hewitt et al., 1976; Misund et al., 1995; Brehmer et al., 2006; Bernasconi et al., 2009; Nishimori et al., 2009). Nishimori et al. (2009) developed a quantitative echo-integration method for estimating fish school abundance from scanning sonars, which showed their clear potential to sample fish in much larger volumes than vertical sounders. Some of the important technical specifications of this class of sonar are listed in Table 5.2.

**Figure 5.21.** Illustration of the combined vertical detection capabilities of the Simrad SH80. In this example, the beams are set to look down 30° from the horizontal plane, while the vertical slice is sounding ahead in line with the ship course (Image from Simrad, Kongsberg Maritime AS).

**Table 5.1. Specifications of three Simrad scanning sonars.**

<table>
<thead>
<tr>
<th>Specification</th>
<th>SH80</th>
<th>SX90</th>
<th>SH90</th>
</tr>
</thead>
<tbody>
<tr>
<td>Frequency (kHz)</td>
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<td>20–30 (1-kHz step)</td>
<td>111.5–116.5 (0.5-kHz step)</td>
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<td>Tilt</td>
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<td>+10 to –60° (0.5° step)</td>
<td>+10 to –60° (0.5° step)</td>
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<td>CW/FM</td>
<td>CW/Chirp</td>
</tr>
<tr>
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<td>256</td>
<td>480</td>
</tr>
<tr>
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<td>Horizontal/vertical (360°/180°)</td>
<td>Horizontal/vertical (360°/180°)</td>
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<tr>
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<td></td>
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<tr>
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<td>360°</td>
<td>360°</td>
<td></td>
</tr>
<tr>
<td>Vertical transmission</td>
<td>7–10.5°</td>
<td>7.5°</td>
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<td></td>
<td>Horizontal</td>
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<td>7.4–11.4°</td>
<td>5.5°</td>
</tr>
<tr>
<td></td>
<td>8°</td>
<td>7.5°</td>
<td></td>
</tr>
</tbody>
</table>

5.6.1 Simrad SH80

This section provides an example of a sphere calibration (Chapter 2) of a Simrad SH80 so the data may be used to quantify backscatter from fish schools (Nottestad et al., 2007) and large whales (Nottestad et al., 2002; Bernasconi et al., 2009, 2011). The SH80 transducer was installed on the ship’s centreline under the bow (Figure 5.22). The sonar samples 64 sectors, each 5.62°, totalling 360°. The axis of each 8°-beamwidth is in the centre of each sector.

Before the calibration, profiles of the environmental conditions were made using a CTD probe. These measurements indicated that the water temperature, $t_w = 15.6°C$, sound speed, $c_w = 1502.8 \text{ m/s}$, and absorption coefficient, $\alpha_a = 35 \text{ dB m}^{-1}$ were stable at the depth of the transducer, $d_w = 6.2 \text{ m}$.

A 75-mm-diameter sphere made from tungsten carbide with 6% cobalt was used to measure the system response for eight 110-kHz beams and pulse durations ranging from 0.8 to 3.8 ms. The sphere weighs 3.3 kg, so is stable when suspended in the water. It has target strength, $TS$, of approximately $–34.4 \text{ dB re 1 m}^2$ at 110 kHz. (A 65-mm-diameter sphere made from tungsten carbide with 6% cobalt may have a more stable $TS$ over the SH80’s range of frequencies).

Consulting the ship’s engineering plans, the approximate positions of each beam axis were marked with tape on the ship’s bow rail. Using these marks as reference, the calibration sphere was lowered to depths coinciding with the beam axis which, with the sonar tilt angle set at 0° (horizontally), was equivalent to the transducer depth (Figure 5.22).
For each of the eight selected beams and pulse duration values, an average of 476 sphere-\( TS \) measurements was recorded. The received voltages, \( u_{er} \) (V) corresponding to backscatter from the sphere, measured at different ranges, were converted from binary to ASCII format (Sonar Data Converter V9.04, Simrad) and echo level, \( EL \) (dB re 1 \( \mu \)Pa) using,

\[
EL = 20 \log_{10}(u_{er}) - M_u
\]

(5.3)

where \( M_u \) is the frequency-dependent sonar receiving sensitivity (\( M_u = -204.8 \) dB re 1 V \( \mu \)Pa\(^{-1} \) at 110 kHz; Simrad; Horten, Norway). The \( EL \) between 5.5 and 7.2 m varied by \(-8.5 \) dB for each beam (average of \(-375 \) pings), similar to variations observed (Otta et al., 2009).

The on-axis gain, \( G_0 \) (dB re 1), was then estimated by solving the sonar equation with values for \( EL \); source level, \( SL \) (dB re 1 \( \mu \)Pa at 1m), the theoretical sphere \( TS \) (dB re 1 m\(^2 \)) (MacLennan, 1981), the range between the transducer and the sphere, \( r \) (m), and the absorption coefficient, \( \alpha_a \) (dB m\(^{-1} \)) (Francois and Garrison, 1982a). The results confirm that \( G_0 \) should be calibrated for each pulse duration used.

The method described above is time consuming and should be refined to expedite the calibration process and more accurately position the sphere on the axis of each beam.
6 Acknowledgements

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7 References


Calibration of acoustic instruments


Author contact information

Lars Andersen
Kongsberg Maritime AS
Kirkegårdsveien 45
Kongsberg, Norway
lars.nonboe.andersen@simrad.com

Laurent Berger
French Research Institute for Exploitation of the Sea
BP 33
La Tremblade 17390, France
Laurent.Berger@ifremer.fr

Matteo Bernasconi
Institute of Marine Research
Nordnesgaten 50
5817 Bergen, Norway
matteo.bernasconi@imr.no

Eckhard Bethke
Thuener Institute of Sea Fisheries
Bundesallee 50
38116 Braunschweig, Germany
eckhard.bethke@ti.bund.de

Kevin Boswell
Florida International University
11200 Southwest 8th Street
Miami, FL 33199, USA
kmboswel@fiu.edu

Dezhang Chu
NOAA Fisheries
1315 East-West Highway
Silver Spring, MD 20910, USA
Dezhang.Chu@noaa.gov

David Demer
NOAA Fisheries
1315 East-West Highway
Silver Spring, MD 20910, USA
David.Demer@noaa.gov

Reka Domokos
NOAA Fisheries
1315 East-West Highway
Silver Spring, MD 20910, USA
Reka.Domokos@noaa.gov
Calibration of acoustic instruments

Adam Dunford
National Institute of Water and Atmospheric Research
41 Market Place
Viaduct Harbour
Auckland Central 1010
Private Bag 99940
Newmarket, Auckland 1149, New Zealand
a.dunford@niwa.co.nz

Sascha Fässler
IMARES Wageningen UR
6708 PB
Wageningen, Netherlands
sascha.fassler@wur.nl

Stephane Gauthier
Department of Fisheries and Oceans
Institute of Ocean Sciences
9860 West Saanich Road
P.O Box 6000
Sidney, British Columbia, V8L 4B2, Canada
Stephane.gauthier@dfo.mpo.gc.ca

Lawrence Hufnagle
NOAA Fisheries
1315 East-West Highway
Silver Spring, MD 20910, USA
Lawrence.C.Hufnagle@noaa.gov

Mike Jech
NOAA Fisheries
1315 East-West Highway
Silver Spring, MD 20910, USA
Michael.Jech@noaa.gov

Naig Le Bouffant
French Research Institute for Exploitation of the Sea
BP 33
La Tremblade 17390, France
Naig.LeBouffant@ifremer.fr

Anne Lebourges-Dhaussy
Research Institute for Development
Le Sextant
44, Bd de Dunkerque, CS 90009
13572 Marseille cedex 02, France
Anne.Lebourges.Dhaussy@ird.fr