

Digital Communication Systems Engineering with Software-Defined Radio

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Lecture 14

Basic Problem

- ▶ We need some sort of way of detecting a pulse transmitted over a channel corrupted by noise
- ▶ Suppose we transmit a pulse signal $g(t)$ through an additive channel with noise $w(t)$, which produces the output $x(t)$:

$$x(t) = g(t) + w(t), \text{ for } 0 \leq t \leq T \quad (1)$$

- ▶ The signal $g(t)$ may represent a “1” or a “0” in a digital communication system
- ▶ The signal $w(t)$ is a sample function of a white noise process of zero mean and power spectral density of $\frac{N_0}{2}$
- ▶ Assume that the receiver has knowledge of all the possible waveforms $g(t)$

Detecting $g(t)$

- ▶ The function of the receiver is to detect the pulse signal $g(t)$ in an optimal manner given the received signal $x(t)$
- ▶ To achieve this, let us filter $x(t)$ in order to minimize the effects of the noise in some statistical sense
 - ▶ We want to enhance the probability of correct detection
- ▶ Suppose the output of the filter $h(t)$ is $y(t) = g_0(t) + n(t)$
 - ▶ Signal $g_0(t)$ is the filtered signal component and $n(t)$ is the filtered noise component
- ▶ Our goal is to maximize $g_0(t)$ with respect to $n(t)$ using peak pulse SNR

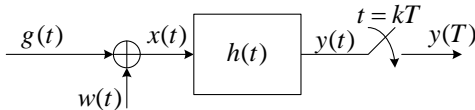


Figure : Filtering Process for Detecting $g(t)$.

Peak Pulse SNR

- ▶ Peak pulse SNR is defined as:

$$\eta = \frac{|g_0(T)|^2}{E\{n^2(t)\}} \quad (2)$$

where $|g_0(T)|^2$ is the instantaneous power of the output signal at sampling instant T , and $E\{n^2(t)\}$ is the average power of the output noise

- ▶ How do we achieve the goal of maximizing $g_0(t)$ relative to $n(t)$?
 - ▶ Design a filter $h(t)$ that achieves the largest possible η

Designing $h(t)$

- ▶ Solving for $h(t)$ consists of evaluating the expression:

$$|g_0(t)|^2 = \left| \int_{-\infty}^{\infty} H(f)G(f)e^{j2\pi ft} df \right|^2, \quad (3)$$

which is the magnitude squared of the inverse Fourier transform of $H(f)G(f) = \mathcal{F}\{h(t) * g(t)\}$

- ▶ Signal $w(t)$ is white with power spectral density $\frac{N_0}{2}$
 - ▶ Power spectral density of $n(t)$ is defined via EWK as $S_N(f) = \frac{N_0}{2}|H(f)|^2$
- ▶ Recall that the mean-square value of a waveform is equal to:

$$E\{n^2(t)\} = R_N(0) = \mathcal{F}^{-1}\{S_N(f)\}|_{\tau=0} = \frac{N_0}{2} \int_{-\infty}^{\infty} |H(f)|^2 df$$

Solving for η

- ▶ Apply the definition for η :

$$\eta = \frac{|g_0(T)|^2}{E\{n^2(t)\}} = \frac{\left| \int_{-\infty}^{\infty} H(f)G(f)e^{j2\pi ft} df \right|^2}{\frac{N_0}{2} \int_{-\infty}^{\infty} |H(f)|^2 df} \quad (4)$$

- ▶ Now we need to solve for $H(f)$ that yields the largest possible η
 - ▶ Use *Schwarz's Inequality* to obtain a closed-form solution

Recall Schwarz's Inequality

- ▶ If we have two complex functions, $\phi_1(x)$ and $\phi_2(x)$, such that:

$$\int_{-\infty}^{\infty} |\phi_1(x)|^2 dx < \infty \quad \text{and} \quad \int_{-\infty}^{\infty} |\phi_2(x)|^2 dx < \infty$$

then:

$$\left| \int_{-\infty}^{\infty} \phi_1(x) \phi_2(x) dx \right|^2 \leq \left(\int_{-\infty}^{\infty} |\phi_1(x)|^2 dx \right) \cdot \left(\int_{-\infty}^{\infty} |\phi_2(x)|^2 dx \right)$$

- ▶ This becomes an equality when $\phi_1(x) = K \cdot \phi_2^*(x)$

Completing the Derivation

- ▶ Applying Schwarz's Inequality to this derivation, we obtain:

$$\left| \int_{-\infty}^{\infty} H(f)G(f)e^{j2\pi ft} df \right|^2 \leq \left(\int_{-\infty}^{\infty} |H(f)|^2 df \right) \cdot \left(\int_{-\infty}^{\infty} |G(f)|^2 df \right)$$

- ▶ Consequently, we get: $\eta \leq \frac{2}{N_0} \int_{-\infty}^{\infty} |G(f)|^2 df$
 - ▶ Thus, we get $H_{\text{opt}}(f) = K \cdot G^*(f)e^{-j2\pi fT}$
- ▶ To determine $H_{\text{opt}}(f)$, solve for the inverse Fourier transform:

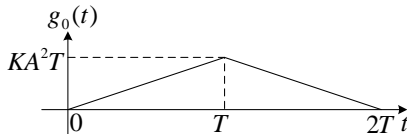
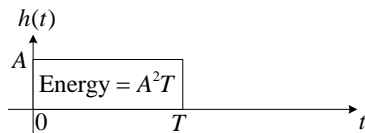
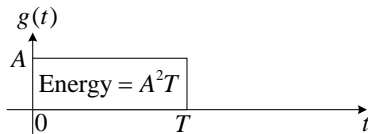
$$h_{\text{opt}}(t) = K \cdot \int_{-\infty}^{\infty} G^*(f)e^{-j2\pi fT} e^{-j2\pi ft} df = K \cdot g(T - t)$$

Why Call It “Matched Filter”?

- ▶ We convolve the time-flipped and time-shifted version of the transmitted pulse with the transmitted pulse itself
- ▶ We are trying to match the pulses→SNR maximizing!!
- ▶ This filter can be implemented using a *tapped delay line*

Matching $h(t)$ with $g(t)$

- ▶ Suppose we have a signal $g(t) \rightarrow$ What is $h(t)$?
 - ▶ Solve $h(t) = g(T - t)$
- ▶ What is the output of $h(t)$?
 - ▶ When we convolve $g(t)$ with $h(t)$, we get a triangular pulse



Matched Filter Realization Schematic

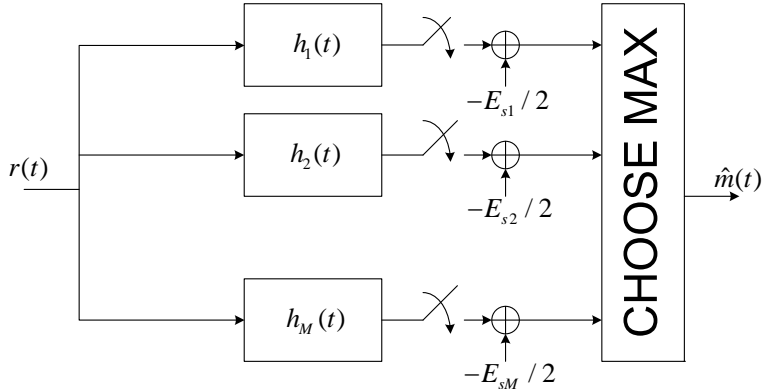


Figure : Schematic of Matched Filter Realization of Receiver Structure.