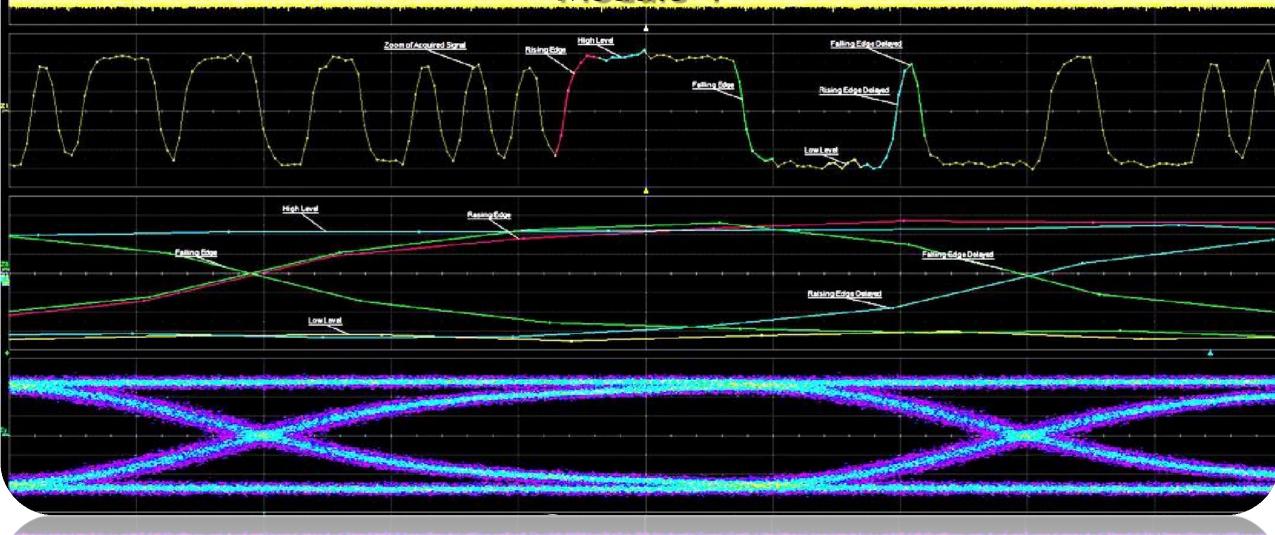
Baseband System Module-4



Topics to be discussed

- Introduction
- Inter Symbol Interference (ISI)
- Role of Matched Filter
- Nyquist criterion for distortion less transmission
- Raised cosine spectrum
- Correlative coding
- Eye pattern
- Equalization

Introduction

• If any rectangular pulse representing one bit of information is applied to the channel input, the shape of the pulse will be distorted at the channel output

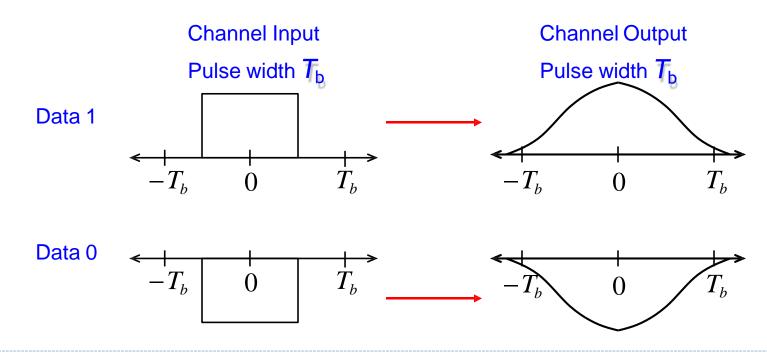
• The distorted pulse may consist of a main lobe representing the original bit of information along with side lobes

• These **side lobes** represent channel distortion also referred as *intersymbol interference*

• Our main objective is that of a *signal design*, whereby the effect of symbol interference is reduced to zero

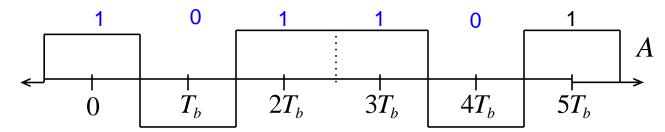
InterSymbol Interference (ISI)

- **Intersymbol interference (ISI)** occurs when a pulse spreads out in such a way that it interferes with adjacent pulses at the sample instant.
- Example: assume polar NRZ line code. The channel outputs are shown as spreaded (width $T_{\rm b}$ becomes $2T_{\rm b}$) pulses shown (Spreading due to bandlimited channel characteristics).

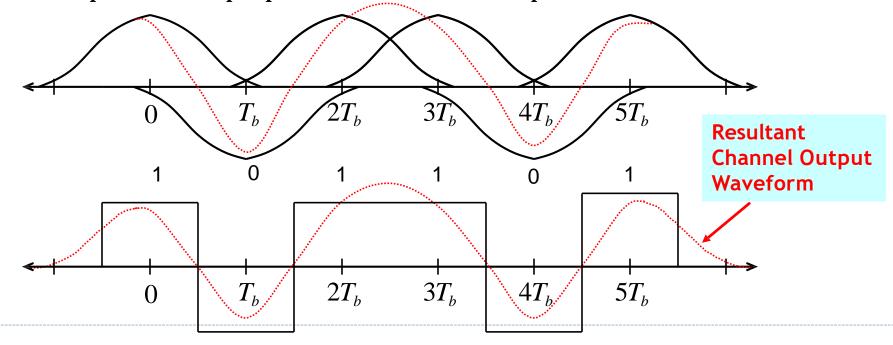


InterSymbol Interference (ISI)

• For the input data stream:

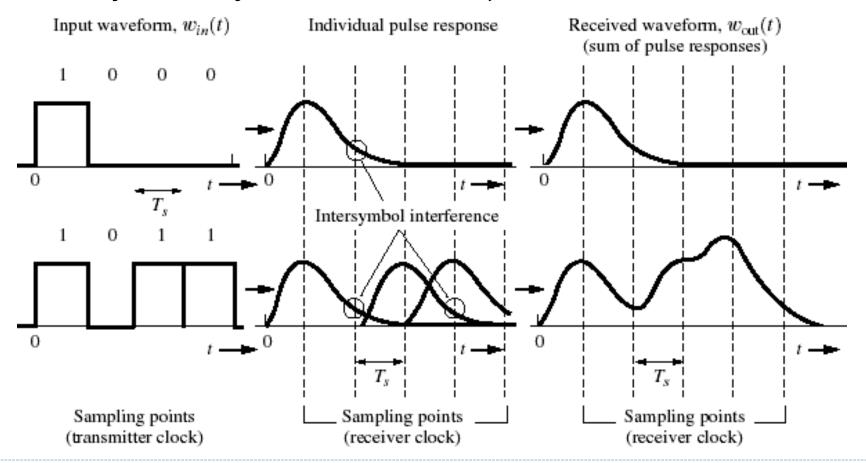


• The channel output is the superposition of each bit's output:



InterSymbol Interference (ISI)

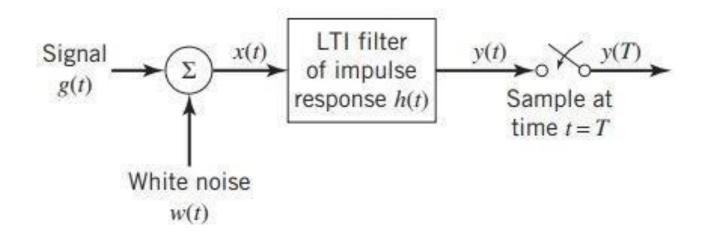
• If the rectangular multilevel pulses are filtered improperly, they will spread in time, and the pulse for each symbol may be smeared into adjacent time slots and cause *ISI*.



- Matched filter is a device for the optimal detection of digital pulse
- The impulse response of the filter is matched to the shape of the input signal
- Matched filter generates an output to maximize the output peak power ratio to mean noise power within its frequency response.
- In telecommunications, it is the optimal linear filter used to increase the SNR or signal-to-noise ratio in the existence of additive stochastic noise.
- The characteristics of matched filter include the following.
- The Signal to noise ratio maximization is possible even for non-Gaussian noise.
- The output generated by this filter is like signal energy in the nonexistence of noise.
- They are applicable for the detection of signals.

Optimum Receiver Filter

• Here, before the transmitted signal g(t) is received at the receiver, it is first passed through a band-limited linear receiver with impulse response h(t) as shown below:



- We assume that the source of uncertainty lies with noise w(t)
- The function of receiver is to detect the signal g(t) in an optimum manner from the received signal x(t), given by:

$$x(t) = g(t) + w(t), \ 0 \le t \le T$$

- Here, g(t) is the transmitted pulse signal which is corrupted by additive white noise process w(t)
- T is an arbitrary observation interval
- Since, the filter is linear, the resulting output y(t) may be expressed as:

$$y(t) = g_0(t) + n(t)$$

- Here, $g_0(t) = g(t) * h(t)$ and n(t) = w(t) * h(t)
- $g_0(t)$ and n(t) are produced by the convolution of signal and noise components of x(t) with impulse response h(t) respectively
- For $g_0(t)$ to be considerably greater than n(t) we maximize the *peak pulse signal-to-noise ratio* given by:

$$\eta = \frac{|g_0(T)|^2}{E[n^2(t)]}$$
, measured at $t = T$

Matched Filter

• Here, we investigate the optimization of impulse response of a filter to maximize the peak signal-to-noise ratio at the output of the filter

• Using Fourier transform, $g_0(t)$ can be expressed in terms of G(f)H(f) as:

$$g_0(t) = \int_{-\infty}^{\infty} H(f)G(f) \exp(j2\pi f t) df$$

• At the sample time t=T, the signal power is given by:

$$|g_0(T)|^2 = \left| \int_{-\infty}^{\infty} H(f)G(f) \exp(j2\pi f T) df \right|^2$$

The power spectral density is given by:

$$S_{NN}(f) = \frac{N_0}{2} |H(f)|^2$$

And, the average output noise power is:

$$E[n^{2}(t)] = \frac{N_{0}}{2} \int_{-\infty}^{\infty} |H(f)|^{2} df$$

Therefore, the peak signal-to-noise ratio will be:

$$\eta = \frac{\left| \int_{-\infty}^{\infty} H(f)G(f) \exp(j2\pi f T) df \right|^2}{\frac{N_0}{2} \int_{-\infty}^{\infty} |H(f)|^2 df}$$

Using Schwartz's inequality, we have:

$$\left| \int_{-\infty}^{\infty} H(f)G(f) \exp(j2\pi f T) df \right|^{2} \le \int_{-\infty}^{\infty} |H(f)|^{2} df \int_{-\infty}^{\infty} G|f|^{2} df$$

The peak signal-to-noise ratio can thus be expressed as:

$$\eta \le \frac{2}{N_0} \int_{-\infty}^{\infty} G|f|^2 df$$

- Correspondingly, H(f) assumes its optimum value denoted by $H_{opt}(f)$
- The optimum value is given by:

$$H_{opt}(f) = kG^*(f)\exp(-j2\pi fT)$$

• Here, k is a scaling factor of appropriate dimensions

 $\mathbb{G}^*(f)$ is the complex conjugate of the Fourier transform of the input signal g(t)

• In terms of time domain, using inverse Fourier transform we have:

$$h_{opt}(t) = kG^*(f) \exp[-j2\pi f(T-t)] df$$

• For a real signal g(t) we have, $G^*(f) = G(-f)$, we may rewrite:

$$h_{opt} = k \int_{-\infty}^{\infty} G(-f) \exp[-j2\pi f(T-t)] df = kg(T-t)$$

• Thus, the impulse response of the optimum filter will be given by:

$$h_{opt}(t) = kg(T - t)$$

• The above equation shows that h_{opt} is the delayed version of the input signal g(t)

• It means that $h_{opt}(t)$ is "matched" to the input signal

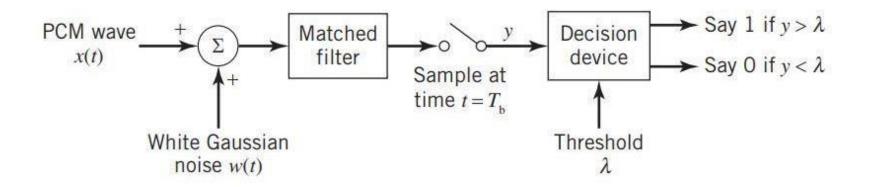
A linear time-invariant filter defined in this way is called a matched filter

- Consider a binary transmission system based on polar nonretum-to-zero (NRZ) signaling
- Here, 0 and 1 are represented by positive and negative rectangular pulses of equal amplitude and duration
- The noise is modeled as additive white Gaussian noise w(t) of zero mean and power spectral density $N_0/2$
- In the signaling interval $0 \le t \le T_b$, the received signal x(t) is given by:

$$x(t) = \begin{cases} +A + w(t), & symbol \ 1 \ was \ sent \\ -A + w(t), & symbol \ 0 \ was \ sent \end{cases}$$

• T_b is the bit duration and A is the transmitted pulse amplitude

- Given the noisy signal x(t), the receiver is required to make a decision in each signalling interval
- Two possible kinds of errors are considered:
- *Error of the first kind*: Symbol 1 is chosen when 0 was actually transmitted
- *Error of the second kind*: Symbol 0 is chosen when 1 was actually transmitted



• The probability density function of the random variable Y, given that symbol 0 was sent

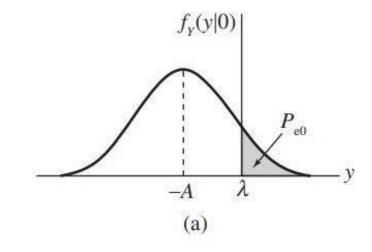
$$f_Y(y|0) = \frac{1}{\sqrt{\pi N_0/T_b}} \exp(-\frac{(y+A)^2}{N_0/T_b})$$

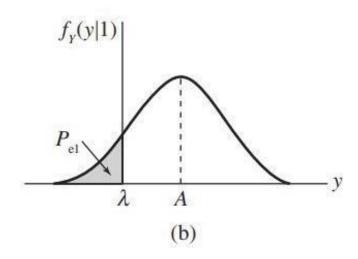
- Let P_{e0} denote the *conditional probability* of error, given that symbol 0 was sent
- This probability is defined by the shaded area under the curve of $f_Y(y|0)$ from the threshold λ to infinity
- When noise is present, y occasionally assumes a value greater than λ , in which case an error is made

Analysis of the effect of channel noise on a binary system

• (a) Probability density function of random variable *Y* at matched filter output when a 0 is transmitted,

• (b) Probability density function of *Y* when a 1 is transmitted





• The probability of this error, conditional on sending 0, is defined by:

$$P_{e0} = P(y > \lambda | symbol\ 0\ was\ sent) = \int_{\lambda}^{\infty} f_Y(y|0) dy = \frac{1}{\sqrt{\pi N_0/T_b}} \int_{\lambda}^{\infty} \exp\left(-\frac{(y+A)^2}{\frac{N_0}{T_b}}\right) dy$$

- Define a new variable: $z = \frac{y+A}{\sqrt{N_0/2T_h}}$
- We may thus reformulate: $P_{e\,0}=\frac{1}{\sqrt{2\pi}}\int_{\sqrt{2E_b/N_0}}^{\infty}\exp\left(-\frac{z^2}{2}\right)dz$
- E_b is the transmitted signal energy per bit defined by $E_b = A^2 T_b$

We now introduce the definition of the so-called Q-function

$$Q(u) = \frac{1}{\sqrt{2\pi}} \int_{u}^{\infty} \exp\left(-\frac{z^2}{2}\right) dz$$

• The conditional probability of error P_{e0} in terms of the Q-function as follows:

$$P_{e\,0} = Q\left(\sqrt{\frac{2E_b}{N_0}}\right)$$

- We also find that $P_{e\,0}=P_{e\,1}$
- A channel for which the conditional error probabilities P_{e1} and P_{e0} are equal is said to be binary symmetric

• The average probability of symbol error P_e in the receiver is given by

$$P_e = p_0 P_{e0} + p_1 P_{e1}$$

- Where, p_0 and p_1 are the *a priori* probabilities of binary symbols 0 and 1, respectively
- Since, $P_{e\,1}=P_{e\,0}$ and $p_0=p_1=\frac{1}{2}$, therefore we finally obtain:

$$P_e = P_{e1} = P_{e0}$$

$$P_e = Q\left(\sqrt{\frac{2E_b}{N_0}}\right)$$

