Digital Communications (ELC 325b)

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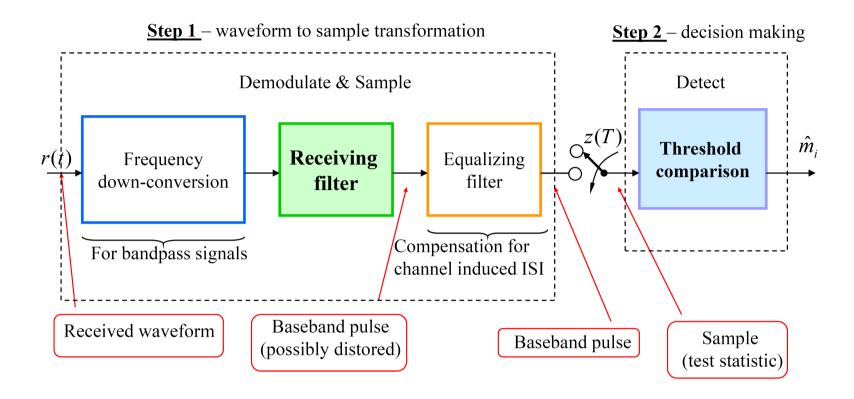
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Outline

- Base-Band Transmission and Structure of Optimum Receiver
 - Inter-Symbol Interference
 - Design of Optimum ISI-Free Communication System

Inter-Symbol Interference



What is ISI?

It is another source of errors that arises when the **communication channel is dispersive**.

It occurs because dispersed symbols are expanded beyond the symbol duration to interfere with adjacent symbols

Inter-Symbol Interference

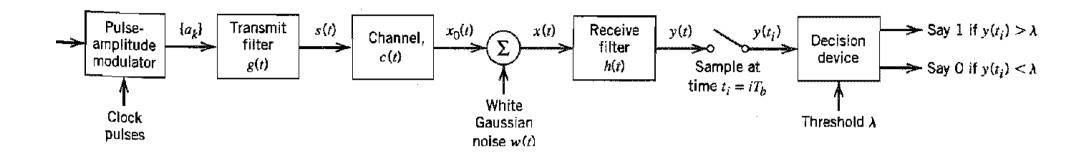
- A rectangular pulse requires an infinite bandwidth on the channel.
 This is cannot be practically achieved
- A band-limited pulse will be widely spread in time (This is refereed to as **Time Spread**)

Requirements of the pulse shape

- Band-limited
- Does not interfere with adjacent pulses

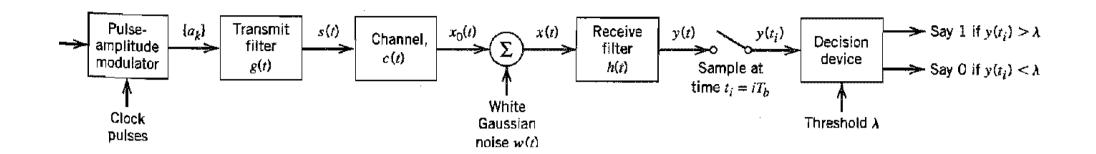
ISI will cause that the amplitudes of the pulses change resulting in less robustness to noise

The analytics of ISI



- ullet The input is binary bits, each of duration T_b
- ullet $\{a_k\}$ is a sequence of amplitude-modulated short pulses. In the case of binary PAM, $a_k=\pm 1$
- s(t) is a sequence of pulse shaped symbols
- y(t) is the output of the receiver filter. It is sampled at $t_i = iT_b$ synchronously with the transmitter
- The decision device finally decides, based on a threshold λ , whether the sample is '1' or '0'

The analytics of ISI



$$s(t) = \sum_{k} a_{k}g(t - kT_{b})$$

$$x_{0}(t) = s(t) * c(t)$$

$$x(t) = x_{0}(t) + w(t) = s(t) * c(t) + w(t)$$

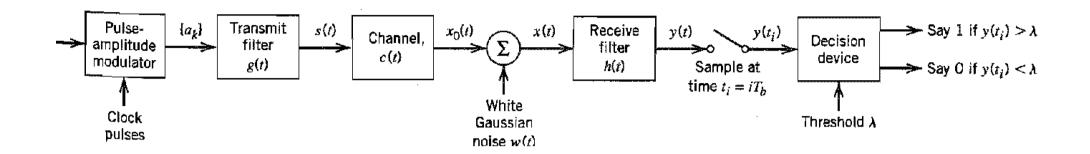
$$y(t) = x(t) * h(t)$$

$$= s(t) * c(t) * h(t) + w(t) * h(t)$$

$$= \mu \sum_{k} a_{k}p(t - kT_{b}) + n(t), \qquad \mu p(t) = g(t) * c(t) * h(t)$$

$$\mu P(f) = G(f)C(f)H(f)$$

The analytics of ISI



$$y(t_i) = \mu \sum_{k} a_k p(t_i - kT_b) + n(t_i)$$
$$= \mu a_i + \mu \sum_{k \neq i} a_k p((i - k)T_b) + n(t_i)$$

- **The first term** represents the contribution of the *i*th bit (This is the bit that needs detection)
- The second term represents the effect of all other transmitted bits on the decoding of the i^{th} bit (This is the ISI)
- The third term is the noise sample at the sampling time

Dealing with ISI

- The presence of ISI and noise in the DCS is unavoidable. This is introduces errors in the decision of the decision device
- In the design of the transmit and receive filters, g(t) and h(t), the objective of minimizing the effects of ISI as well as noise should be considered

Note that ideally $y(t_i) = \mu a_i$

Problem Statement

How to design of the optimum p(t) = g(t) * c(t) * h(t), such that effects of ISI and noise are minimized?

Design of ISI-Free Systems

For the system to be ISI-free

$$p(nT_b) = p((i-k)T_b) = \begin{cases} 1, & i=k \\ 0, & i \neq k \end{cases}$$

Note: Such condition results in perfect detection of the transmitted symbols, in the absence of noise.

 $p(nT_b)$ are samples of p(t), sampled by a train of impulses, then using the sampling theorem,

$$P_s(f) = \frac{1}{T_b} \sum_{-\infty}^{\infty} P\left(f - n \frac{1}{T_b}\right)$$

where $P_s(f)$ is the Fourier Transform of $p(nT_b)$ Since $p(nT_b) = \delta(n)$, then $P_s(f) = 1$

Nyquist's Criterion

Nyquist's Criterion

The condition of zero ISI is

$$\sum_{-\infty}^{\infty} P(f - nR_b) = T_b, \quad \text{where } R_b = \frac{1}{T_b}$$

Nyquist's criterion for distortion-less baseband transmission in the absence of noise states that the frequency function P(f) will eliminate the ISI for samples taken at intervals of T_b provided that the above equation is satisfied

Note: The RHS of the Nyquist's Criterion need not be exactly T_b . It can be any constant.

Simplest way to satisfy the Nyquist's Criterion

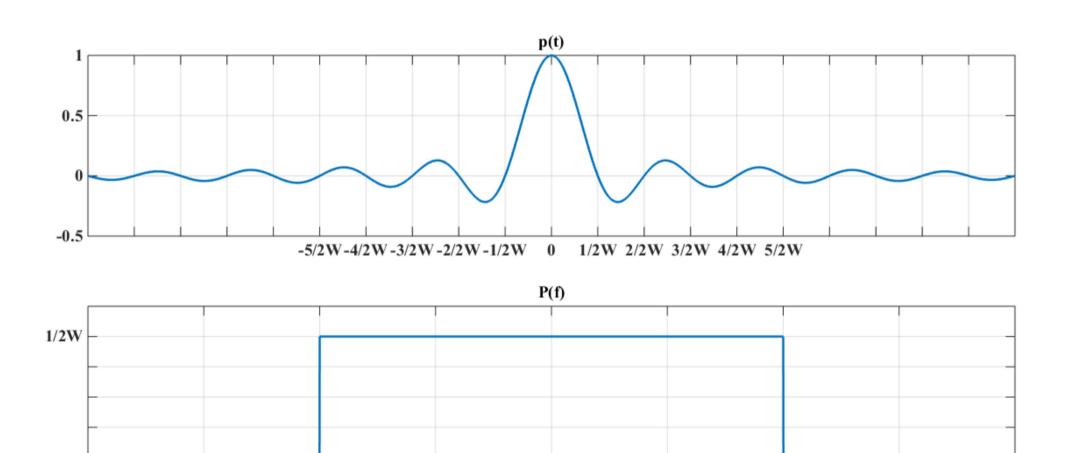
Ideal Nyquist Channel: Rectangular Form

$$P(f) = \frac{1}{2W} \operatorname{rect}\left(\frac{f}{2W}\right) = \begin{cases} \frac{1}{2W}, & -W < f < W \\ 0, & |f| > W \end{cases}$$

$$p(t) = \operatorname{sinc}(2Wt) = \frac{\sin(2\pi Wt)}{2\pi Wt}$$

Nyquist B.W.
$$W=\frac{R_b}{2}=\frac{1}{2T_b}$$

Nyquist Rate $R_b=2W$
 $T_b=\frac{1}{2W}$



 \mathbf{W}

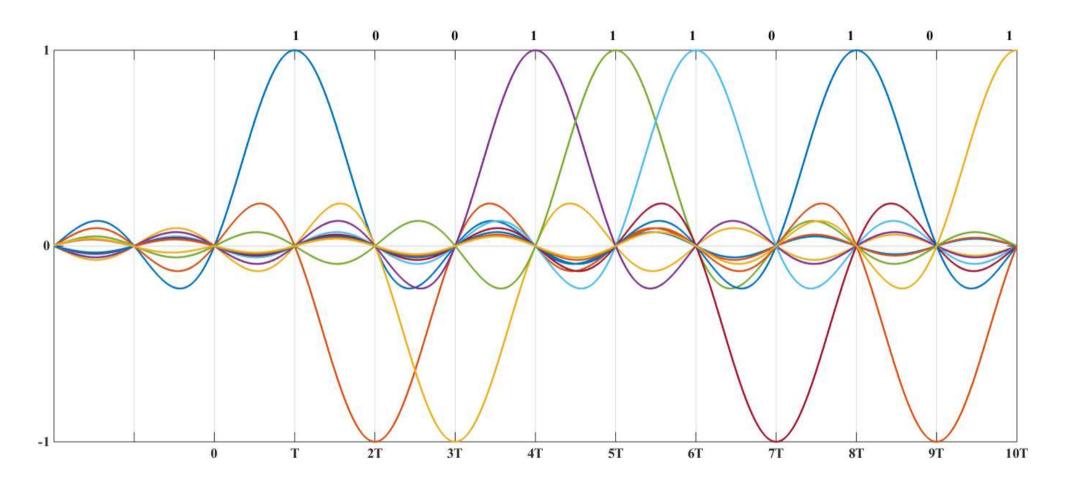
 $-\mathbf{W}$

0 └ -2W

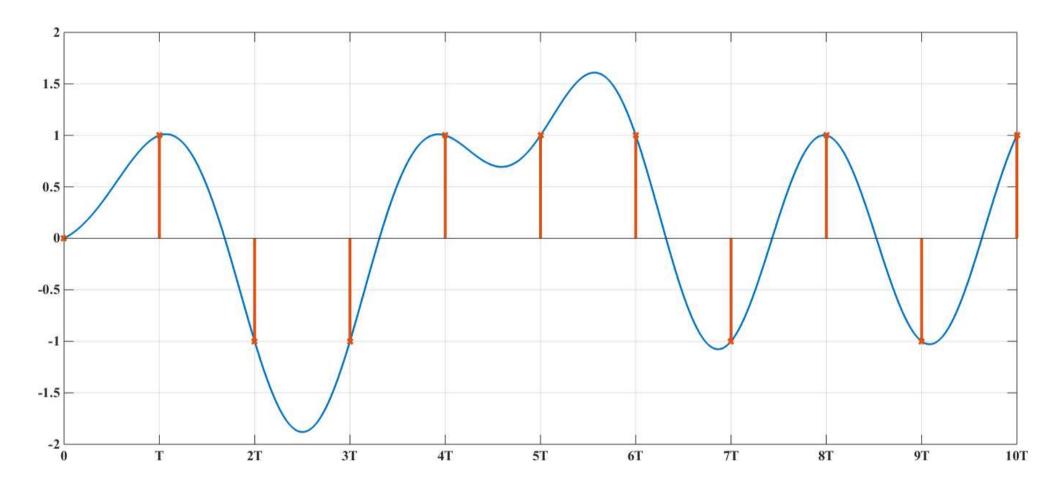
0

2W

Transmitted Stream $= 1 \ 0 \ 0 \ 1 \ 1 \ 1 \ 0 \ 1 \ 1$



Transmitted Stream = $1\ 0\ 0\ 1\ 1\ 1\ 0\ 1\ 0$

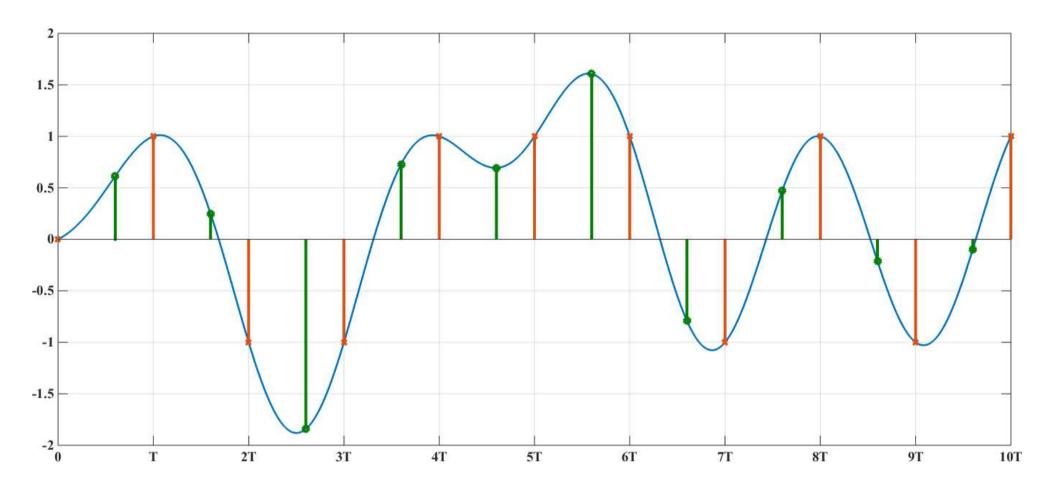


Sampling at $t_i = iT$ Detected Stream = 1 0 0 1 1 1 0 1 0 1

Advantages and Disadvantages

- Economic bandwidth usage: Solves the problem of ISI using the minimum possible bandwidth
- There are practical difficulties:
 - The sudden transition at $f = \pm W$ is physically unrealizable
 - p(t) decreases as $\frac{1}{|t|}$ resulting in a slow rate of decay. So, if the sampling times are slightly shifted, large ISI (because of many previous and following pulse signals) will be caused.

Transmitted Stream $= 1 \ 0 \ 0 \ 1 \ 1 \ 1 \ 0 \ 1 \ 1$



Sampling at $t_i = iT - \epsilon$ Detected Stream = 1 1 0 1 1 1 0 1 0 0 Effect of Sync = $\sqrt{X} \sqrt{\sqrt{\sqrt{A}} \sqrt{X}}$



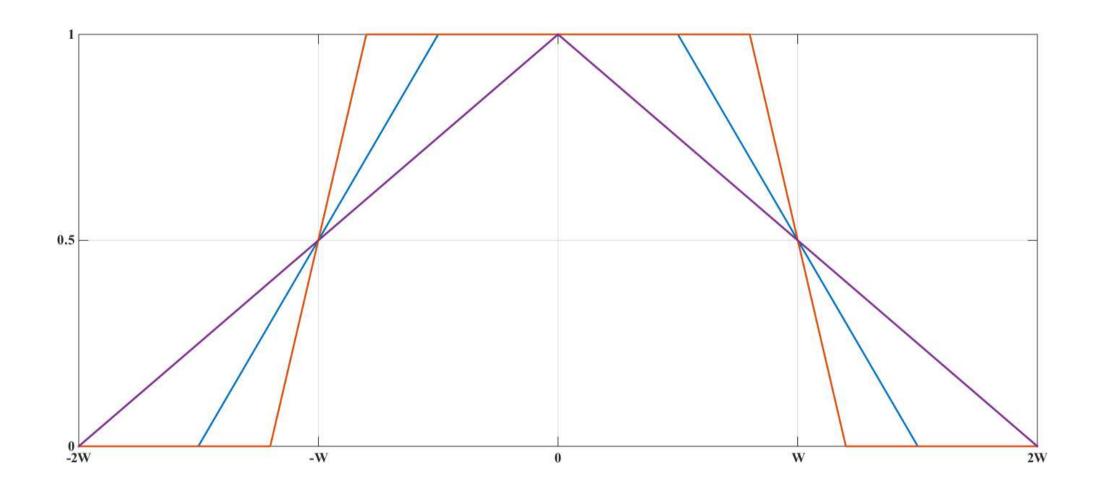
Reduced Nyquist's Criterion

In order to overcome the practical difficulties arising from using the ideal Nyquist channel, we can extend the bandwidth from its minimum value, $W = \frac{R_b}{2}$ to an adjustable value between W and 2W, such that

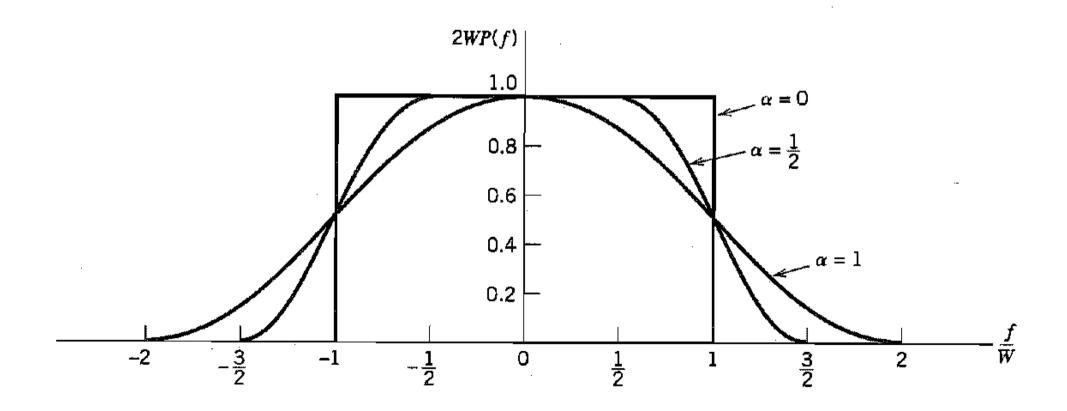
$$P(f) + P(f - 2W) + P(f + 2W) = \frac{1}{2W}, \quad -W \le f \le W$$

Many band-limited functions that satisfy the above equation can be found.

Reduced Nyquist's Criterion



One of the most common forms of P(f) is the **Raised Cosine Spectrum**, which consists of a **flat portion** and a **roll-off portion** of a sinusoidal form.



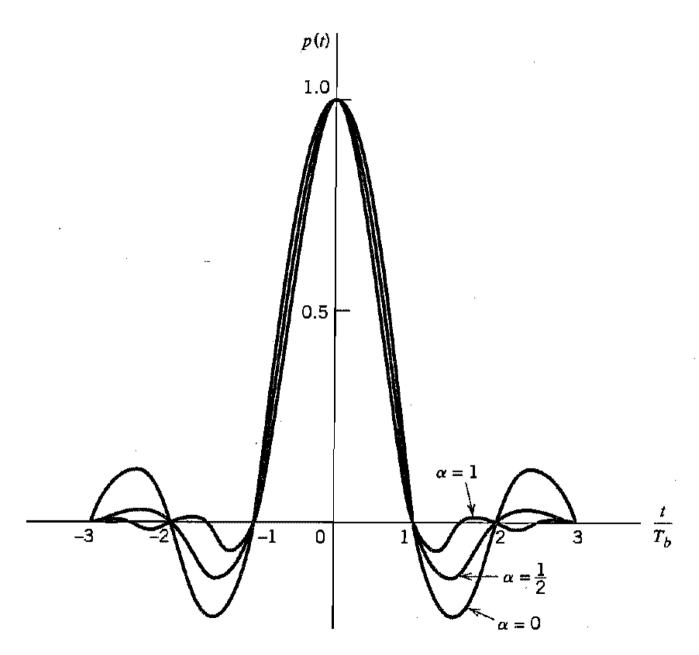
Raised Cosine Spectrum

$$P(f) = \begin{cases} \frac{1}{2W}, & 0 \le |f| \le (1 - \alpha)W \\ \frac{1}{4W} \left[1 - \sin\left(\frac{\pi(|f| - W)}{2\alpha W}\right) \right], & (1 - \alpha)W \le |f| \le (1 + \alpha)W \\ 0, & (1 + \alpha)W \le |f| \end{cases}$$

$$p(t) = sinc(2Wt) \frac{\cos(2\pi\alpha Wt)}{1 - 16\alpha^2 W^2 t^2}$$

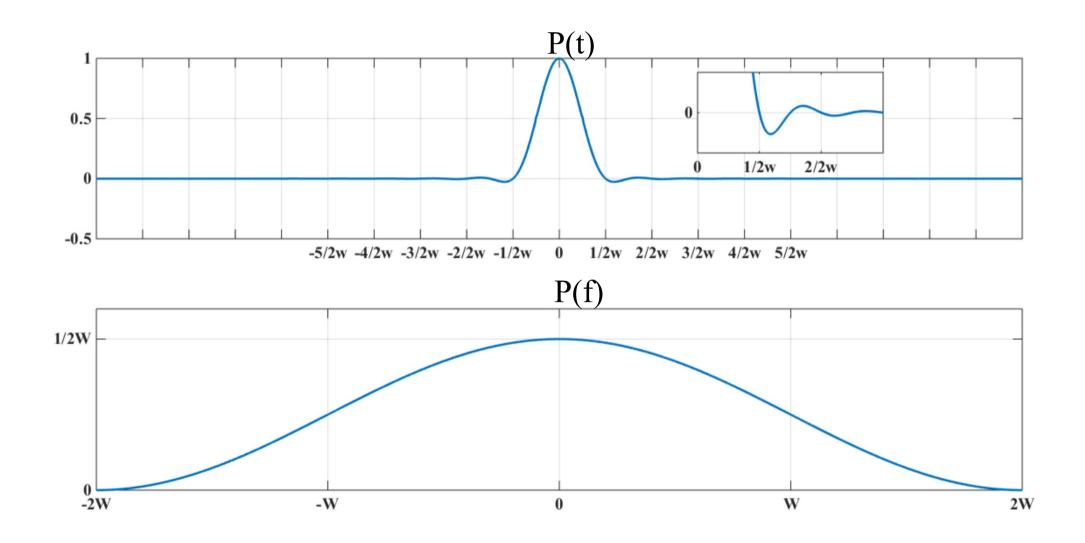
$$f_1 = (1-lpha)W$$
Roll-off Factor $lpha = 1-rac{f_1}{W}$
 $0 \le lpha \le 1$

Transmission B.W.
$$B_T = 2W - f_1 = (1 + \alpha)W$$

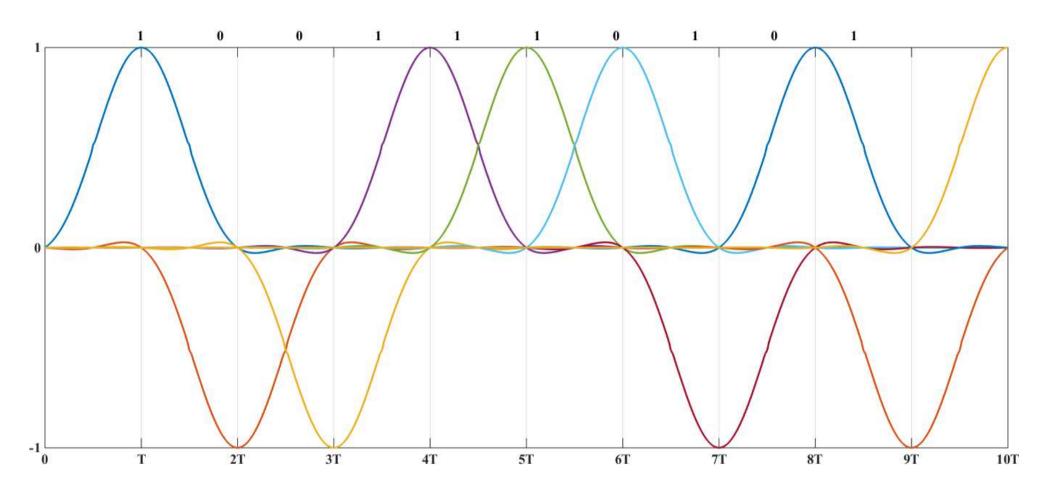


Features of the Raised Cosine Spectrum

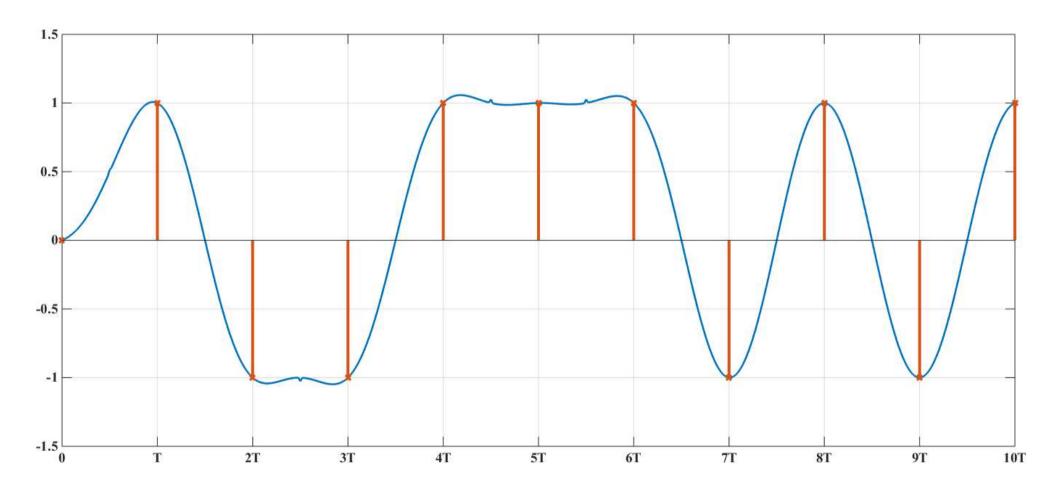
- \bullet P(f): Gradual cut-off
- 2 p(t): Sinc component \rightarrow zero-crossings at $t_i = iT \rightarrow \mathsf{ISI}\text{-Free}$
- 3 p(t): Fast decay $\frac{1}{|t|^2}$
- For $\alpha = 1$: Full Roll-Off Larger B.W. Extra Zero-crossings



Transmitted Stream $= 1 \ 0 \ 0 \ 1 \ 1 \ 1 \ 0 \ 1 \ 1$



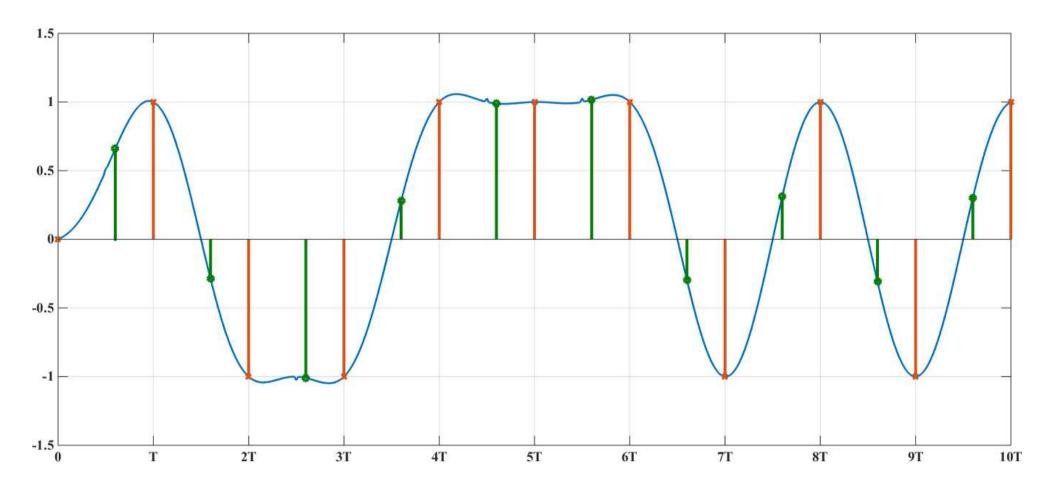
Transmitted Stream $= 1 \ 0 \ 0 \ 1 \ 1 \ 1 \ 0 \ 1 \ 1$



Sampling at $t_i = iT$ Detected Stream = 1 0 0 1 1 1 0 1 0 1

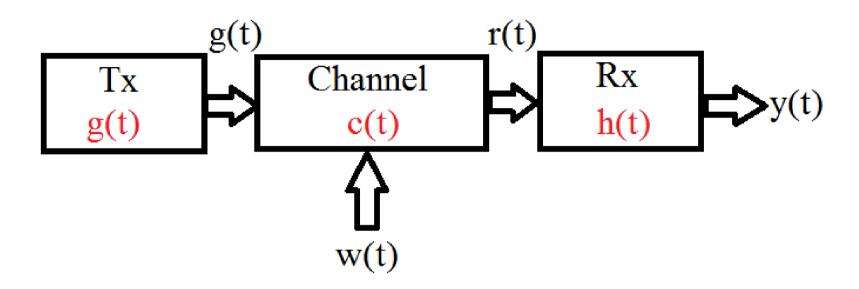


Transmitted Stream $= 1 \ 0 \ 0 \ 1 \ 1 \ 1 \ 0 \ 1 \ 0 \ 1$





Design of Optimum ISI-Free Communication System



Recall:

$$p(t) = g(t) * c(t) * h(t)$$

$$P(f) = G(f)C(f)H(f)$$

Conditions to satisfy:

- Condition of Matched Filter
- Condition of ISI-Free



Design of Optimum ISI-Free Communication System

Condition of Matched Filter

$$H(f) = G^*(f)e^{-j2\pi fT}$$

The exponential term represents a phase shift, equivalent to time delay.

Condition of ISI-Free

$$H(f)G(f) = P(f)$$

It is assumed that the channel is flat, C(f) = const., for at least the maximum possible B.W. of the pulse P(f).

$$G(f) = \sqrt{P(f)}$$

$$H(f) = \sqrt{P(f)}$$

Square-Root Raised Cosine Spectrum

Transmission Bandwidth

Assuming a symbol duration of T,

$$B_T = (1+\alpha)W$$

$$= (1+\alpha)\frac{R_{\mathrm{S}}}{2} = (1+\alpha)\frac{1}{2T_{\mathrm{S}}}$$

In the case of Binary Transmission,

$$T_{\rm S} = T_b$$

$$B_T = (1+\alpha)\frac{1}{2T_b}$$

In the case of M-ary Transmission,

$$T_s = (\log_2 M) T_b$$

$$B_T = (1 + \alpha) \frac{1}{2T_s}$$

$$= (1 + \alpha) \frac{1}{\log_2 M} \frac{1}{2T_b}$$

Transmission Bandwidth

Example

A computer puts out binary data at 56 kbps. The computer output is transmitted using a baseband binary PAM system that has a raised-cosine spectrum. Determine the transmission bandwidth for $\alpha=0.25, 0.5, 0.75, 1$. Repeat if each of 3 successive binary digits are coded into one of eight PAM levels.

$$B_T = (1 + \alpha)W, \qquad W = \frac{R_b}{2} = 28 \text{ kbps}$$

$$B_T = (1+\alpha)W, \qquad W = \frac{R_s}{2} = \frac{\frac{R_b}{\log_2 8}}{2} = \frac{28}{3} \text{ kbps}$$

References



Simon Haykin (2001) Communication Systems, 4th Edition. *John Wiley*.

Thank You

Questions?