

S-72.245 Transmission Methods in Telecommunication Systems (4 cr)

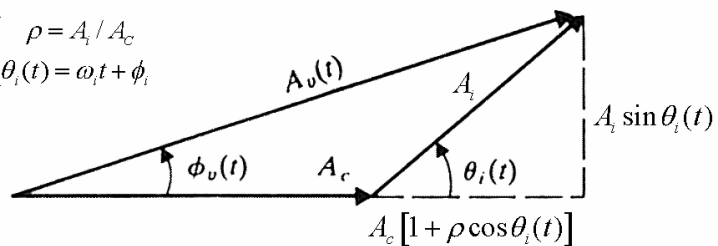
Review

Additive interference in unmodulated carrier

- Consider a general tone interference signal

$$v(t) = \underbrace{A_c \cos(\omega_c t)}_{\text{carrier}} + \underbrace{A_i \cos[(\omega_c + \omega_i)t + \phi_i]}_{\text{interference}}$$

$$\begin{cases} \rho = A_i / A_c \\ \theta_i(t) = \omega_i t + \phi_i \end{cases}$$



- interference produces both AM and FM:

$$A_v(t) = A_c \sqrt{1 + \rho^2 + 2\rho \cos \theta_i(t)}$$

$$\phi_v(t) = \arctan \frac{\rho \sin \theta_i(t)}{1 + \rho \cos \theta_i(t)}$$

Additive interference and demodulators

- Further simplification under weak interference: $A_i \ll A_c, \rho \ll 1$

$$A_v(t) = A_c \sqrt{1 + \underbrace{\rho^2}_{\approx \rho^2 \cos^2 \theta_i(t)} + 2\rho \cos \theta_i(t)} \approx A_c [1 + \rho \cos \theta_i(t)]$$

$$\phi_v(t) = \arctan \frac{\rho \sin \theta_i(t)}{1 + \rho \cos \theta_i(t)} \approx \arctan [\rho \sin \theta_i(t)] \approx \rho \sin \theta_i(t)$$

- Demodulation functions:

$$y_D(t) \approx \begin{cases} K_{D,AM} A_v(t), \text{ AM} \\ K_{D,PM} \phi(t), \text{ PM} \\ K_{D,FM} d\phi(t) / dt, \text{ FM} \end{cases}$$

- And therefore

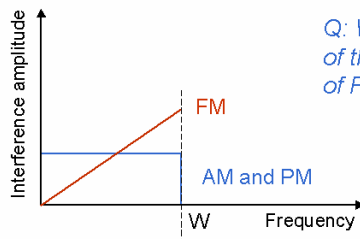
$$y_D(t) \approx \begin{cases} K_{D,AM} A_c [1 + \rho \cos \theta_i(t)], \text{ AM} \\ K_{D,PM} \rho \sin \theta_i(t), \text{ PM} \\ K_{D,FM} \rho f_i \cos \theta_i(t), \text{ FM} \end{cases}$$
- $d[\rho \sin \theta_i(t)] / dt$
 $= d[\rho \sin(\omega_i t + \phi_i)] / dt$
 $= \rho \omega_i \cos(\omega_i t + \phi_i)$

Implications for demodulator design

$$y_D(t) \approx \begin{cases} K_{D,AM} A_c [1 + \rho \cos \theta_i(t)], \text{ AM} \\ K_{D,PM} \rho \sin \theta_i(t), \text{ PM} \\ K_{D,FM} \rho f_i \cos \theta_i(t), \text{ FM} \end{cases}$$

$$\begin{cases} \rho = A_i / A_c \\ \theta_i(t) = \omega_i t + \phi_i \end{cases}$$

- In AM and PM a tone interference produces a tone to reception whose amplitude is comparable to ρ and position comparable to $\theta_i(t) = \omega_i t + \phi_i$
- Interference in FM is more severe the more remote the interfering tone is from the carrier (but still at the reception band W):

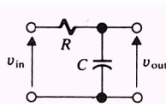


Q: What are the implications of this to noise sensitivity of FM bandwidth determination

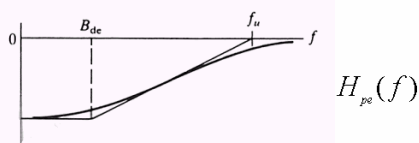
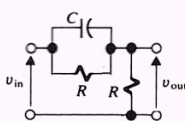
FM preemphases and deemphases filters

- FM related noise emphases can be suppressed by *pre-distortion* and post detection filters (preemphases and deemphases filters):

receiver filter



transmitter filter



Q: What would happen if the filters would be reversed? (TX filter in receiver & vice versa)

$$H_{de}(f) = [1 + j(f / B_{de})]^{-1} \approx \begin{cases} 1, & |f| \ll B_{de} \\ B_{de} / (jf), & |f| \gg B_{de} \end{cases} \text{LPF}$$

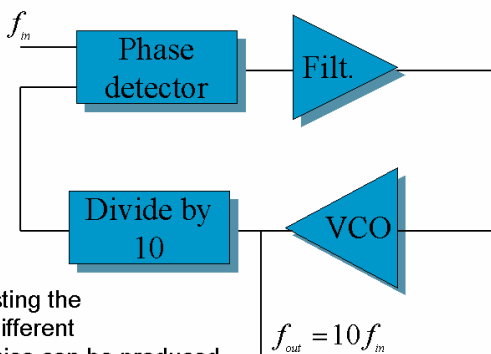
$$H_{pe}(f) = [1 + j(f / B_{de})] \approx \begin{cases} 1, & |f| \ll B_{de} \\ j(f / B_{de}), & |f| \gg B_{de} \end{cases} \text{HPF}$$

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PLL based frequency synthesizer

Reference signal f_m is locked for instance to the fundamental frequency of a crystal oscillator



By adjusting the divider different frequencies can be produced whose phase is locked into f_m

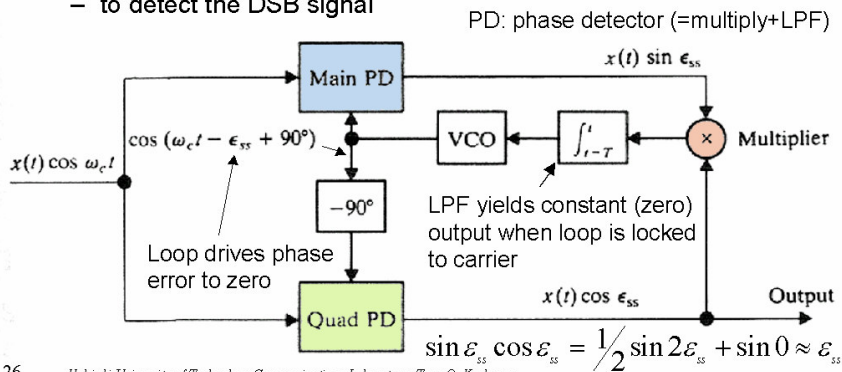
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Detecting DSB using PLL-principle

- An important application for PLLs is in **synchronization** of receiver local oscillator in synchronous detection
- In the **Costas PLL** (below) two phase discriminators are used to:
 - cancel out DSB modulation $x(t)$ in the driving signal
 - synchronize the output frequency to the center frequency of the DSB spectra (the suppressed carrier)
 - to detect the DSB signal

Costas PLL detector for DSB



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PAM Power Spectral Density (PSD)

- PSD for PAM can be determined by using a general expression
Amplitude autocorrelation

$$G_x(f) = \frac{1}{D} |P(f)|^2 \sum_{n=-\infty}^{\infty} R_a(n) \exp(-j2\pi n f D)$$

- For uncorrelated message bits

$$R_a(n) = \begin{cases} \sigma_a^2 + m_a^2, & n = 0 & \text{Total power} \\ m_a^2, & n \neq 0 & \text{DC power} \end{cases}$$

and therefore

$$\sum_{n=-\infty}^{\infty} R_a(n) \exp(-2\pi n f D) = \sigma_a^2 + m_a^2 \sum_{n=-\infty}^{\infty} \exp(-j2\pi n f D)$$

on the other hand $\sum_{n=-\infty}^{\infty} \exp(-j2\pi n f D) = \frac{1}{D} \sum_{n=-\infty}^{\infty} \delta\left(f - \frac{n}{D}\right)$ and $r = 1/D$

$$G_x(f) = \sigma_a^2 r |P(f)|^2 + m_a^2 r^2 \sum_{n=-\infty}^{\infty} |P(nr)|^2 \delta(f - nr)$$

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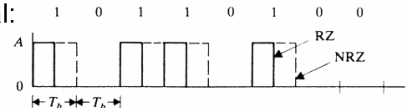
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Example

- For unipolar binary RZ signal:

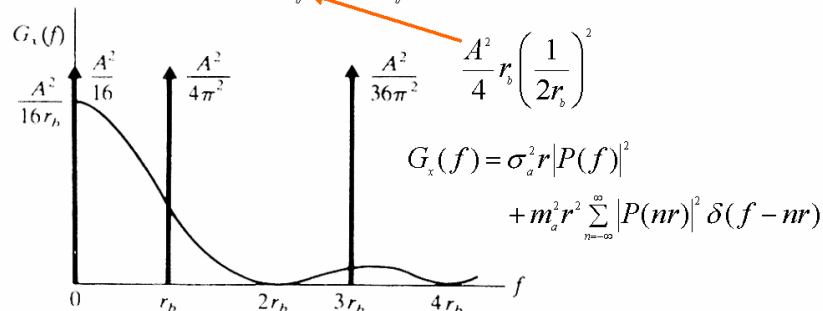
$$P(f) = \frac{1}{2r_b} \operatorname{sinc} \frac{f}{2r_b}$$



- Assume source bits are equally alike and independent, thus

$$\sigma_a^2 = (1/2T_b) \int_0^{T_b/2} A^2 dt = A^2/4, m_a^2 = \sigma_a^2$$

$$\Rightarrow G_x(f) = \frac{A^2}{16r_b} \operatorname{sinc}^2 \frac{f}{2r_b} + \frac{A^2}{16} \sum_{n=-\infty}^{\infty} \delta(f - nr_b) \operatorname{sinc}^2 \frac{n}{2}$$

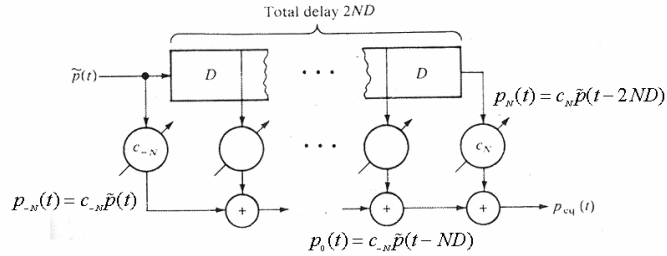


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Equalization: Removing Residual ISI

- Consider a tapped delay line equalizer with



- Search for the tap gains c_N such that the output equals zero at sample intervals D except at the decision instant when it should be unity. The output is (think for instance paths c_{-N} , c_N or c_0)

$$p_{eq}(t) = \sum_{n=-N}^M c_n \tilde{p}(t - nD - ND)$$

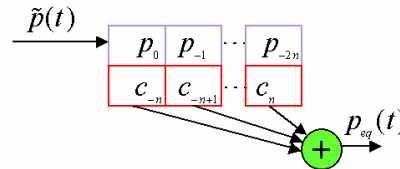
that is sampled at $t_k = kD + ND$ yielding

$$p_{eq}(kD + ND) = \sum_{n=-N}^M c_n \tilde{p}(kD - nD) = \sum_{n=-N}^M c_n \tilde{p}[D(k - n)]$$

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Tapped Delay Line: Matrix Representation



- At the instant of decision:

$$p_{eq}(t_k) = \sum_{n=-N}^N c_n \tilde{p}[D(k-n)] = \sum_{n=-N}^N c_n \tilde{p}_{k-n} = \begin{cases} 1, & k=0 \\ 0, & k=\pm 1, \pm 2, \dots, \pm N \end{cases}$$

- That leads into $(2N+1) \times (2N+1)$ matrix where $(2N+1)$ tap coefficients can be solved:

$$\begin{aligned} \tilde{p}_0 c_{-N} + \tilde{p}_{-1} c_{-N+1} + \dots + \tilde{p}_{-2N} c_N &= 0 \\ \tilde{p}_1 c_{-N} + \tilde{p}_0 c_{-N+1} + \dots + \tilde{p}_{-2N+1} c_N &= 0 \\ &\dots \\ \tilde{p}_N c_{-N} + \tilde{p}_{N-1} c_{-N+1} + \dots + \tilde{p}_{-N} c_N &= 1 \\ &\dots \\ \tilde{p}_{2N} c_{-N} + \tilde{p}_{2N-1} c_{-N+1} + \dots + \tilde{p}_0 c_N &= 0 \end{aligned}$$

$$\begin{bmatrix} \tilde{p}_0 & \dots & \tilde{p}_{-2N} \\ \vdots & \dots & \vdots \\ \tilde{p}_{N-1} & \dots & \tilde{p}_{-N-1} \\ \tilde{p}_N & \dots & \tilde{p}_{-N} \\ \tilde{p}_{N+1} & \dots & \tilde{p}_{-N+1} \\ \vdots & \dots & \vdots \\ \tilde{p}_{2N} & \dots & \tilde{p}_0 \end{bmatrix} \begin{bmatrix} c_{-N} \\ \vdots \\ c_{-1} \\ c_0 \\ c_1 \\ \vdots \\ c_N \end{bmatrix} = \begin{bmatrix} 0 \\ \vdots \\ 0 \\ 1 \\ 0 \\ \vdots \\ 0 \end{bmatrix}$$

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Example of Equalization

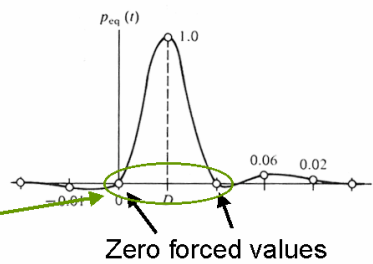
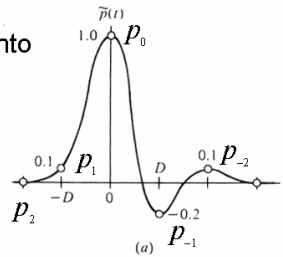
- Read the distorted pulse values into matrix from fig. (a)

$$\begin{bmatrix} 1.0 & 0.1 & 0.0 \\ -0.2 & 1.0 & 0.1 \\ 0.1 & -0.2 & 1.0 \end{bmatrix} \begin{bmatrix} c_{-1} \\ c_0 \\ c_1 \end{bmatrix} = \begin{bmatrix} 0 \\ 1 \\ 0 \end{bmatrix}$$

and the solution is

$$\begin{bmatrix} c_{-1} \\ c_0 \\ c_1 \end{bmatrix} = \begin{bmatrix} -0.096 \\ 0.96 \\ 0.2 \end{bmatrix}$$

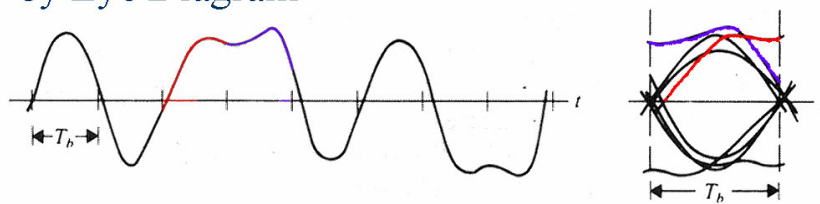
Question: what does these zeros help because they don't exist at the sampling instant?



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Monitoring Transmission Quality by Eye Diagram



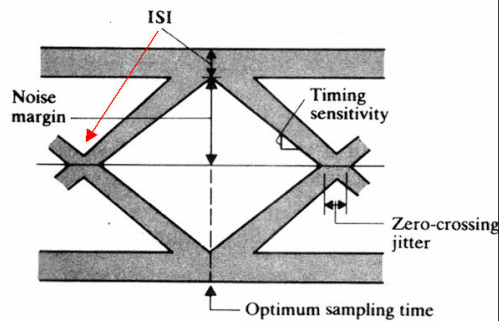
Required minimum bandwidth is

$$B_T \geq r/2$$

Nyquist's sampling theorem:

Given an ideal LPF with the bandwidth B it is possible to transmit independent symbols at the rate:

$$B_T \geq r/2 = 1/(2T_b)$$



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