

# UESTC4004

## Digital Communications

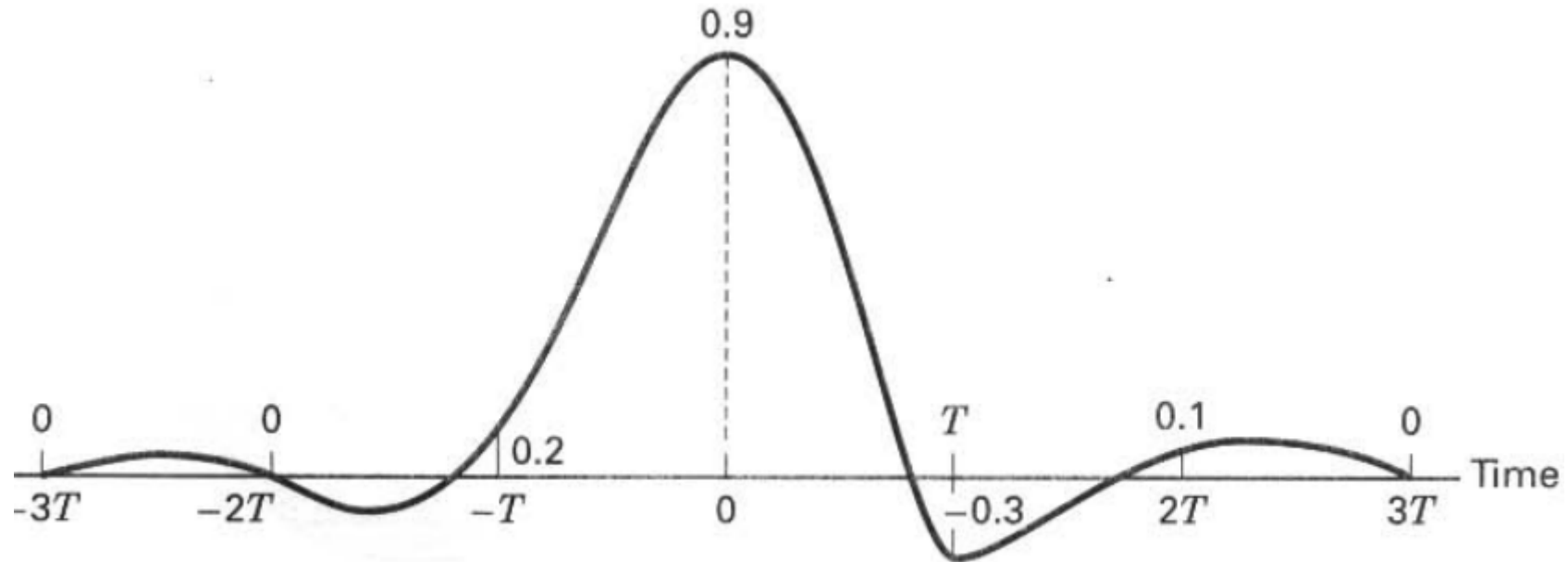
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## 3.4 Equalization

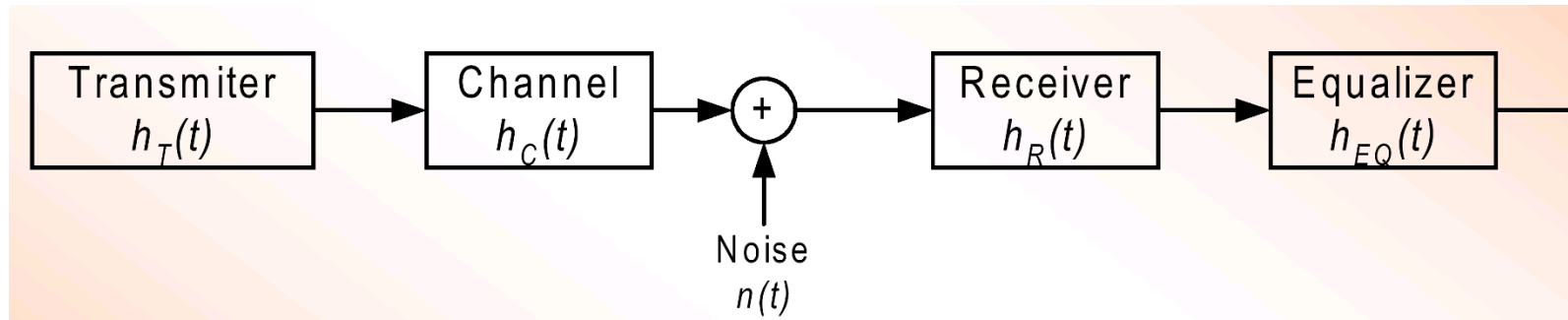
- Nyquist filtering and pulse shaping schemes assumes that the channel is precisely known and its characteristics do not change with time
- However, in practice we encounter channels whose frequency response are either unknown or change with time
  - For example, each time we dial a telephone number, the communication channel will be different because the communication route will be different
  - However, when we make a connection, the channel becomes time-invariant
  - The characteristics of such channels are not known a priori
- Examples of time-varying channels are radio channels
  - These channels are characterized by time-varying frequency response characteristics

# Channel distortions



**Figure 3.25** Received pulse exhibiting distortion.

- To compensate for channel induced ISI we use a process known as **Equalization**: a technique of correcting the frequency response of the channel
- The filter used to perform such a process is called an **equalizer**



- Since  $H_R(f)$  is matched to  $H_T(f)$ , we usually worry about  $H_C(f)$
- The goal is to pick the frequency response  $H_{EQ}(f)$  of the **equalizer** such that

$$H_c(f)H_{EQ}(f) = 1 \Rightarrow H_{EQ}(f) = \frac{1}{H_C(f)} e^{-j\theta_c(f)}$$

- where

- $|H_{EQ}(f)| = \frac{1}{|H_C(f)|}$  and the phase characteristics  $\Theta_{EQ}(f) = -\Theta_C(f)$

# Problems with Equalization

- It can be difficult to determine the inverse of the channel response
  - If the channel response is zero at any frequency, then the inverse is not defined at that frequency
  - The receiver generally does not know what the channel response is. Channel changes in real time so equalization must be adaptive
- The equalizer can have an infinite impulse response even if the channel has a finite impulse response
  - The impulse response of the equalizer must usually be truncated

# Equalization Techniques or Structures

## ■ **Three Basic Equalization Structures**

### □ **Linear Transversal Filter**

- Simple implementation using Tap Delay Line or FIR filters
- FIR filter has guaranteed stability (although adaptive algorithm which determines coefficients may still be unstable)

### □ **Decision Feedback Equalizer**

- Extra step in subtracting estimated residual error from signal

### □ **Maximal Likelihood Sequence Estimator (Viterbi)**

- “Optimal” performance
- High complexity and implementation problem (not heavily used)

# Linear Transversal Equalizer

- This is simply a linear filter with adjustable parameters
- The parameters are adjusted on the basis of the measurement of the channel characteristics
- A common choice for implementation is the **transversal filter** (Tap Delay Line) or the FIR filter with adjustable tap coefficient

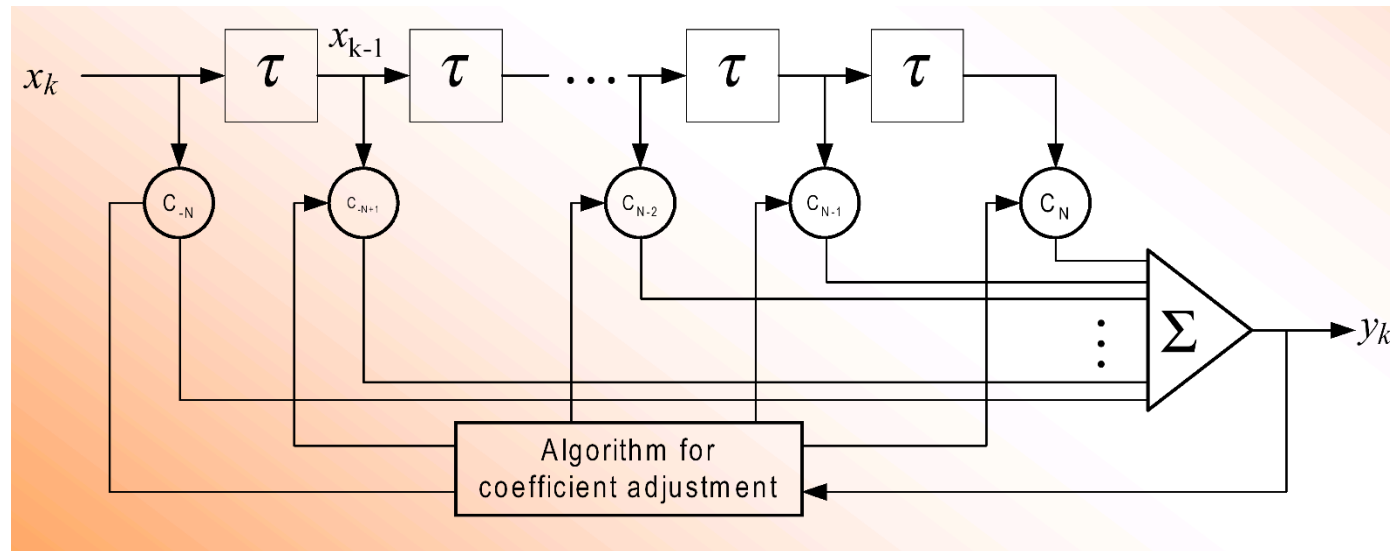


Fig. 3.26

- **Total number of taps =  $2N+1$**
- **Total delay =  $2N\tau$**

- $N$  is chosen sufficiently large so that equalizer spans length of the ISI.
- Normally the ISI is assumed to be limited to a finite number of samples
- The output  $y_k$  of the Tap Delay Line equalizer in response to the input sequence  $\{x_k\}$  is

$$y_k = \sum_{n=-N}^N x(k-n)c_n, \quad k = -2N, \dots, 2N$$

where  $c_n$  is the weight of the  $n^{\text{th}}$  tap

- Ideally, we would like the equalizer to eliminate ISI resulting in

$$y_k = \begin{cases} 1, & k = 0 \\ 0, & k \neq 0 \end{cases}$$

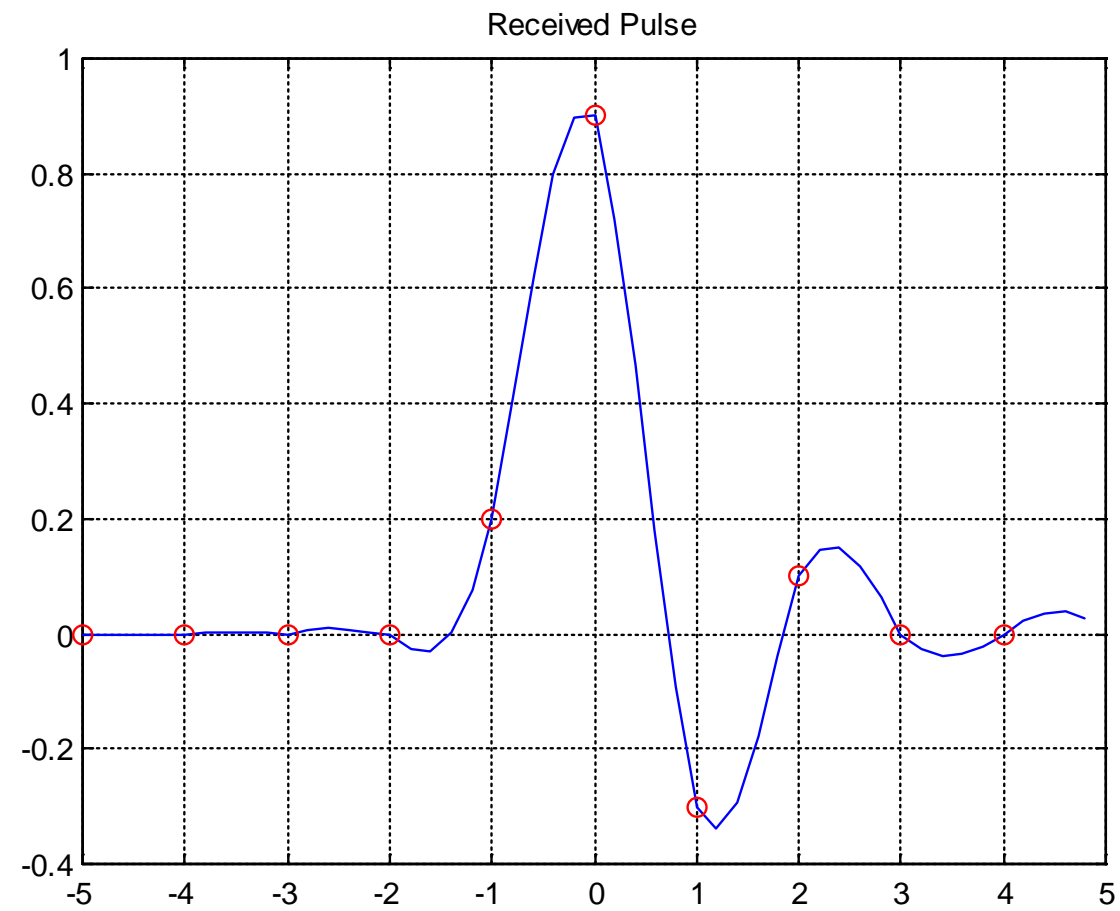
- But this cannot be achieved in practice.



- Zero forcing Equalizer : However, the tap gains can be chosen such that

$$y_k = \begin{cases} 1, & k = 0 \\ 0, & k = \pm 1, \pm 2, \dots, \pm N \end{cases}$$

### Example 3.5: (Page 155)



# Zero-Forcing Solution

- For  $N=1$

$$k = -1, \quad y(-1) = c_{-1}x(-1 - (-1)) + c_0x(-1 - (0)) + c_1x(-1 - (1))$$

$$k = 0, \quad y(0) = c_{-1}x(0 - (-1)) + c_0x(0 - (0)) + c_1x(0 - (1))$$

$$k = 1, \quad y(1) = c_{-1}x(1 - (-1)) + c_0x(1 - (0)) + c_1(1 - (1))$$

$$\Rightarrow \begin{bmatrix} y(-1) \\ y(0) \\ y(1) \end{bmatrix} = \begin{bmatrix} x(0) & x(-1) & x(-2) \\ x(1) & x(0) & x(-1) \\ x(2) & x(1) & x(0) \end{bmatrix} \begin{bmatrix} c_{-1} \\ c_0 \\ c_1 \end{bmatrix}$$

$(2N + 1) \times (2N + 1) \qquad (2N + 1) \times 1$

- For N=2

$$\begin{bmatrix} y(-2) \\ y(-1) \\ y(0) \\ y(1) \\ y(2) \end{bmatrix} = \begin{bmatrix} x(0) & x(-1) & x(-2) & x(-3) & x(-4) \\ x(1) & x(0) & x(-1) & x(-2) & x(-3) \\ x(2) & x(1) & x(0) & x(-1) & x(-2) \\ x(3) & x(2) & x(1) & x(0) & x(-1) \\ x(4) & x(3) & x(2) & x(1) & x(0) \end{bmatrix} \begin{bmatrix} c_{-2} \\ c_{-1} \\ c_0 \\ c_1 \\ c_2 \end{bmatrix}$$

Generalizing results:

$$\mathbf{c} = \mathbf{X}^{-1} \mathbf{y}$$

$$\text{where } \mathbf{y} = \begin{bmatrix} 0 \\ 1 \\ 0 \end{bmatrix} \text{ for } N = 1$$

## Example

- Design a 3-taps zero forcing equalizer for input  $x(n) = \{0, -0.1, 0.15, 0.87, 0.12, -0.2, 0\}$  in which  $x(0) = 0.87$ .
- Answer:  $c=[0.236 \ 1.220 \ -0.226]'$ .