

# UESTC4004 Digital Communications

Equalization



#### **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 apriori
- Examples of time-varying channels are radio channels
  - These channels are characterized by time-varying frequency response characteristics

### Channel distortions

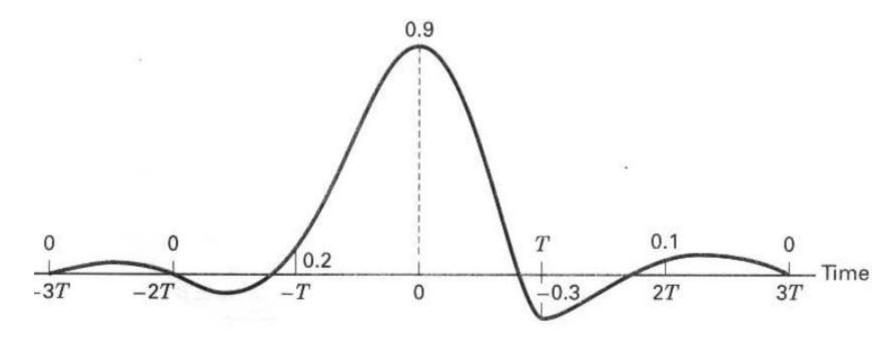
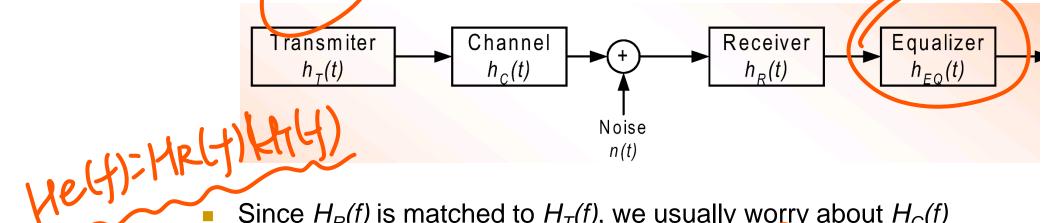


Figure 3.25 Received pulse exhibiting distortion.



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  $H_{FQ}(f)$  of the equalizer

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

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

$$\Theta_{EQ}(f) = -\Theta_{C}(f)$$

Note that  $H_c(f) = |H_c(f)|e^{j\theta_c(f)}$ 



#### **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**

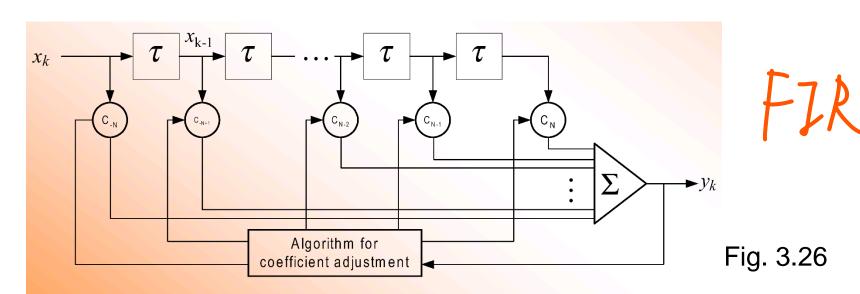
#### Three Basic Equalization Structures

- 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



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

3taps = 2N+1 > N=



- 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

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

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

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

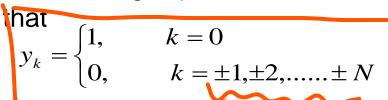
$$y_k = \begin{cases} 1, & k = 0 \\ 0, & k \neq 0 \end{cases}$$

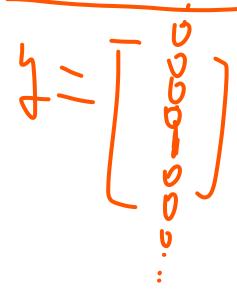
But this cannot be achieved in practice.





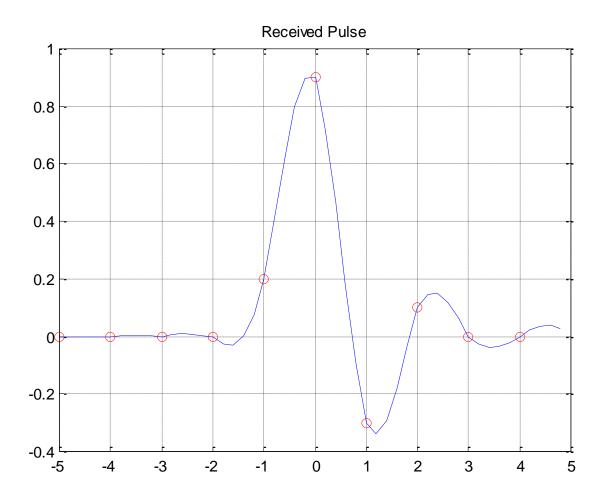
Zero forcing Equalizer: However, the tap gains can be chosen such







#### **Example 3.5: (Page 155)**





#### **Zero-Forcing Solution**

For N=1

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

$$k = -1, \quad y(-1) = c_{-1}x(-1 - (-1)) + c_0x(-1 - (0)) + c_1x(-1 - (1))$$
 
$$k = 0, \quad y(0) = c_{-1}x(0 - (-1)) + c_0x(0 - (0)) + c_1x(0 - (1))$$
 
$$k = 1, \quad y(1) = c_{-1}x(1 - (-1)) + c_0x(1 - (0)) + c_1(1 - (1))$$

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

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

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



#### For N=2







 $c_{-2}$ 

 $C_{-1}$ 

 $C_1$ 

$$y(-2)$$
 $y(-1)$ 
 $y(0)$ 
 $y(1)$ 
 $y(2)$ 

$$\begin{bmatrix} x(0) & x(-1) & x(-2) & x(-3) & x(-4) \\ x(1) & x(0) & x(-1) & x(-2) & x(-3) \\ x(2) & x(1) & x(0) & x(-1) & x(-2) \\ x(3) & x(2) & x(1) & x(0) & x(-1) \\ x(4) & x(3) & x(2) & x(1) & x(0) \end{bmatrix}$$

Generalizing results

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



where 
$$y = \begin{bmatrix} 0 \\ 1 \\ 0 \end{bmatrix}$$
 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 -0.2226]'.

#### Minimum MSE Solution

- The tap weights  $c_n$  are chosen to minimize the mean-square error (MSE) of all ISI terms plus noise power at the output of the equalizer
  - ullet Set of overdetermined equations is used to obtain a minimum MSE solution by multiplying both sides of the equation by  $X^T$
  - - X is a non-square matrix with dimension 4N + 1 by 2N + 1
    - y and c are vectors with 4N + 1 and 2N + 1, respectively
  - $\square \quad R_{Xy} = R_{XX}c$ 
    - $\mathbf{R}_{\mathbf{X}\mathbf{y}} = \mathbf{X}^T \mathbf{y}$  is called the cross-correlation vector
    - $\mathbf{R}_{XX} = \mathbf{X}^T \mathbf{X}$  is called the auto-correlation vector
  - The tap weight can be obtained as
    - $c = R_{XX}^{-1} R_{Xy}$
- Compared with zero-forcing MSE solution is more robust in the presence of noise and large ISI



## Review Questions

Given a received distorted set of pulse samples {x(k)}, with voltage values 0.0, 0.2, 0.9, -0.3, 0.1 as shown in slide 10, design a 3 tap zero-forcing equalizer to reduce the ISI.

[-0.2140 0.9631 0.3448]

Given that the channel transfer function H(f)=1/(1+j2πf). Derive the expression for the corresponding equalizer transfer function and sketch both its amplitude and phase characteristics

$$|A| = |A| = |A|$$