ECE 463: Digital Communications Lab.

Lecture 6: Channel Estimation & Correction Haitham Hassanieh



Previous Lecture:

- ✓ ASK Modulation
- ✓ FSK Modulation (Coherent & Non-Coherent)
- ✓ Frame Synchronization

This Lecture:

- Multipath Channel
- Channel Estimation & Correction
- Narrowband vs. Wideband Channels
- Channel Equalization

Digital Communication System



The Channel



Channel:

- Adds Noise
- Attenuates the Signal
- Rotates the Phase of the Signal
- Delays the Signal

$$h\propto \frac{\lambda}{d'}e^{j2\pi d'/\lambda}$$





Send Training Sequence (Preamble Bits): Known Bits

$$x(0) = 1 \longrightarrow y(0) = h + v(0)$$

$$x(1) = 1 \longrightarrow y(1) = h + n(1)$$

$$x(2) = -1 \longrightarrow y(2) = -h + n(2)$$

Estimate channel: $\tilde{h} = \sum_{k} \frac{y(k)}{x(k)}$ Correct channel: $\tilde{x}(t) = \frac{y(t)}{\tilde{h}}$

The Channel





Multipath Propagation: radio signal reflects off objects ground, arriving at destination at slightly different times

$$y(t) = \alpha_1 e^{\phi_1} x(t - \tau_1) + \alpha_2 e^{\phi_2} x(t - \tau_2) + \alpha_3 e^{\phi_3} x(t - \tau_3) \cdots$$

$$y(t) = \sum_{k} \alpha_{k} e^{\phi_{k}} x(t - \tau_{k}) = \sum_{k} h(\tau_{k}) x(t - \tau_{k}) = h(t) * x(t)$$

h(t) is channel impulse response.



Symbols arriving along late paths interfere with following symbols.



paths sum up destructively

h(t) is channel impulse response.

$$y(t) = \sum_{k} h(\tau_k) x(t - \tau_k) = h(t) * x(t)$$

Channel Fading: Symbols arriving along different paths sum up destructively

Example 2 paths with distance $d_1 = 1m$, $d_2 = 1.06m$:

$$h = h_1 + h_2 = \frac{\lambda}{d_1} e^{j2\pi d_1/\lambda} + \frac{\lambda}{d_2} e^{j2\pi d_2/\lambda}$$

$$= \frac{\lambda}{d_1} e^{j2\pi d_1/\lambda} \left(1 + \frac{d_1}{d_2} e^{j2\pi (d_2 - d_1)/\lambda} \right) \qquad \frac{d_1}{d_2} \approx 1$$

$$if \ \frac{d_2 - d_1}{\lambda} \approx \frac{1}{2} \to \ h = \frac{\lambda}{d_1} \ e^{j2\pi d_1/\lambda} (1 + e^{j\pi}) = 0 \qquad \text{Destructive Interference!}$$

h(t) is channel impulse response.

$$y(t) = \sum_{k} h(\tau_k) x(t - \tau_k) = h(t) * x(t)$$

Channel Fading: Symbols arriving along different paths sum up destructively

Example 2 paths with distance $d_1 = 1m$, $d_2 = 1.06m$:

$$h = h_{1} + h_{2} = \frac{\lambda}{d_{1}} e^{j2\pi d_{1}/\lambda} + \frac{\lambda}{d_{2}} e^{j2\pi d_{2}/\lambda}$$

@f_{1} = 2.5GHz (λ = 12 cm): $h = 0.12 e^{j\frac{2\pi}{3}} + 0.113 e^{j\frac{5\pi}{3}} \approx 0.006$
@f_{2} = 5GHz (λ = 6 cm): $h = 0.06 e^{j\frac{5\pi}{3}} + 0.05 e^{j\frac{5\pi}{3}} \approx 0.116$
T7×
Frequency Selective Fading (24dB)

h(t) is channel impulse response.



Multi-tap Channel

h(t) is channel impulse response.



Symbols arriving along late paths

interfere with following symbols.

Problematic in Wideband Channel!







h(t) is channel impulse response.



h(t) is channel impulse response.



h(t) is channel impulse response.



Narrowband Channel is Approximated by a Single Tap Channel

h(t) is channel impulse response.



Narrowband Channel is Approximated by a Single Tap Channel

Narrowband vs. Wideband Channel

Narrowband Channel

h x(t)

Wideband Channel

h(t) * x(t)



Narrowband vs. Wideband Channel



Need to correct for ISI to be able to decode correctly!

Inter-Symbol-Interference

Sources of ISI:

• Multi-tap Channel

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

• Pulse Shaping

$$y(t) = h(t) * p(t) * s(t)$$

• Other hardware filters

How to deal with ISI? Channel Equalization

Channel Equalization



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

$$\hat{x}(t) = f(t) * y(t) = f(t) * h(t) * x(t)$$

Ideally, $f(t) = h^{-1}(t)$

$$\hat{x}(t) = h^{-1}(t) * h(t) * x(t) = \delta(t) * x(t) = x(t)$$



Can Estimate h(t) as an FIR Filter:

- Limited number of paths
- Longer path power decays quickly
- Estimate as a filter with L taps
- Send a training sequence!

Send a training sequence of length N:

$$y[n] = \sum_{l=0}^{L} h[l]t[n-l] + v[n]$$

We will use Least Squares Estimator:

$$\{\hat{h}[0], \hat{h}[1], \dots, \hat{h}[L]\} = \operatorname{argmin}_{h[0], h[1], \dots, h[L]} \sum_{n=L}^{N-1} \left| y[n] - \sum_{l=0}^{L} h[l]t[n-l] \right|^2$$

Send a training sequence of length N:

$$y[n] = \sum_{l=0}^{L} h[l]t[n-l] + v[n]$$



$$\mathbf{y} = \mathbf{A}\mathbf{h} + \mathbf{v}$$

Send a training sequence of length N:

$$y[n] = \sum_{l=0}^{L} h[l]t[n-l] + v[n]$$

$$\mathbf{y} = \mathbf{A}\mathbf{h} + \mathbf{v}$$

To solve for h, we need: $N - L \ge L + 1 \rightarrow N \ge 2L + 1$

Least Squares Solution: minimize
$$\|\mathbf{y} - \mathbf{Ah}\|^2$$

 $\mathbf{\hat{h}} = (\mathbf{A}^{\dagger}\mathbf{A})^{-1}\mathbf{A}^{\dagger}\mathbf{y}$

A[†]: Hermitian, Conjugate Transpose

Channel Equalization



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

$$\hat{x}(t) = f(t) * y(t) = f(t) * h(t) * x(t)$$

Ideally, $f(t) = h^{-1}(t)$

$$\hat{x}(t) = h^{-1}(t) * h(t) * x(t) = \delta(t) * x(t) = x(t)$$

Correcting for h(t)

$$y[n] = \sum_{l=0}^{L} h[l]x[n-l] + v[n]$$

Need to find an inverse filter f, such that:

$$\sum_{l=0}^{L'} f[l]\hat{h}[n-l] = \delta[n-d]$$

- **h** is FIR, h^{-1} is IIR \rightarrow Hard to satisfy exactly.
- *d* is equalization delay
- Try different d to find the best equalizer

Correcting for h(t)

$$y[n] = \sum_{l=0}^{L} h[l]x[n-l] + v[n]$$

Need to find an inverse filter f, such that:

$$\sum_{l=0}^{L'} f[l]\hat{h}[n-l] = \delta[n-d] \quad \underline{L'} \ge \underline{L}$$

\widehat{\mathbf{H}}: Toeplitz Matrix

Channel Equalization



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

$$\hat{x}(t) = f(t) * y(t) = f(t) * h(t) * x(t)$$

Ideally, $f(t) = h^{-1}(t)$

$$\hat{x}(t) = h^{-1}(t) * h(t) * x(t) = \delta(t) * x(t) = x(t)$$

Channel Equalization

Estimating Correcting h(t)for h(t)Solve Least Solve Least Squares Squares $\hat{\mathbf{h}} = \left(\mathbf{A}^{\dagger}\mathbf{A}\right)^{-1}\mathbf{A}^{\dagger}\mathbf{y}$ $\hat{\mathbf{f}} = \left(\widehat{\mathbf{H}}^{\dagger}\widehat{\mathbf{H}}\right)^{-1}\widehat{\mathbf{H}}^{\dagger}\mathbf{\delta}_{\mathbf{d}}$ Apply f(t) to $\gamma(t)$

Need to Solve Least Squares Twice! Computationally Intensive! Channel Equalization Direct Least Squares Skip Estimating h altogether

• Send a training sequence of length N:

$$y[n] = \sum_{l=0}^{L} h[l]t[n-l] + v[n]$$

• Find inverse filter f such that:

$$\sum_{l=0}^{L'} f[l]y[n-l] = t[n-d] \to t[n] = \sum_{l=0}^{L'} f[l]y[n+d-l]$$

 $\mathbf{t} = \mathbf{Y}\mathbf{f}$ \longrightarrow Least Squares: $\hat{\mathbf{f}} = (\mathbf{Y}^{\dagger}\mathbf{Y})^{-1}\mathbf{Y}^{\dagger}\mathbf{t}$



Channel Equalization Eliminating ISI vs Amplifying Noise

ZF (Zero Forcing) Equalizer:

 $y(t) = h(t) * x(t) + v(t) \quad \Leftrightarrow \quad Y(f) = H(f)X(f) + V(f)$

$$\hat{X}(f) = \frac{Y(f)}{H(f)} = X(f) + \frac{V(f)}{H(f)}$$
Noise Amplification
No ISI
Channel Attenuation: $H(f) < 1$
Channel Fading: $H(f) \approx 0$

Good Equalizer balances between reducing ISI & amplifying noise

Channel Equalization		
Computational Complexity vs Convergence Time		
Algorithms to Solve Least Squares:		
N: Length of training sequence		
	Computation	Convergence
LMS: Least Mean Squares	2 <i>N</i> + 1	> 10 <i>N</i>
RLS: Recursive Least Squares	$2.5N^2 + 4.5N$	$\approx N$
Larger <i>N</i> : • More computation		

- Longer training sequences
- Lower error: ISI & Noise

Definitions & Variables

- x(t): Transmitted Signal
- v(t): Additive Gaussian Noise
- y(t): Received Signal
- τ: Time delay of the signal
- *h*: Single Tap Channel Coefficient.
- \tilde{h} : Estimate of the Channel Coefficient.
- au_k : Time delay of the k^{th} propagation path
- α_k : Attenuation of the k^{th} propagation path
- ϕ_k : Phase of the k^{th} propagation path
- h(t): Multi-Tap Channel Impulse Response
- H(f): Frequency Response of the Channel
- *X*(*f*): Frequency Spectrum of transmitted signal
- *V*(*f*): Frequency Spectrum of noise
- *Y*(*f*): Frequency Spectrum of received signal
- h[l]: Coefficient of the l^{th} channel tap
- $\hat{h}[l]$: Estimate of the channel coefficients

- T: Symbol time
- * : Convolution
- λ : Wavelength of the signal.
- *d'*: Distance between TX and RX
- f_c : Carrier Frequency
- p(t): Pulse of pulse shaping filter
- *s*(*t*): Modulated Symbols
- $h^{-1}(t)$: Inverse Channel Response
- f(t): Channel Equalization Filter
- $\hat{x}(t)$: Equalized Signal
- $\delta(t)$: Impulse Function
- y[n]: Sampled received signal
- t[n]: Training Sequence
- *N*: Length of the Training Sequence
- f[l]: Equalization Filter coefficients
- *n*: Symbol index

- **y**: Vector of y[n] samples
- **h**: Vector of *h*[*l*] coefficients
- A: Matrix of t[n] training samples
- **v**: Vector of v[n] noise samples
- **f**: Vector oxf *f*[*l*] coefficients
- **f**: Estimate of **f**
- **h**: Estimate of **h**
- \hat{f} : Estimate of f
- \widehat{H} : Toeplitz matrix of h
- **t**: Vector of t[n] training samples
- Y: Matrix of y[n] received samples
- *d*: Equalization delay
- ()⁻¹: Matrix Inverse
- ()[†]: Conjugate Transpose of Matrix
- *L*: Number of channel taps
- *L*': Number of Equalization filter taps