



Lecture 2
Channel Equalization

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Lecture 2: Channel Equalization 2 Advanced Digital Communications (EQ2410)¹

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Wednesday, Jan. 20, 2016
10:00-12:00, B24

¹Textbook: U. Madhow, *Fundamentals of Digital Communications*, 2008

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Notes



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Overview

Lecture 1

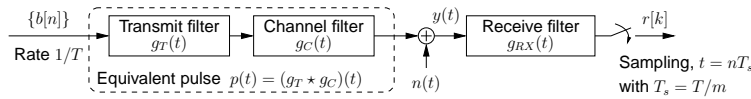
- Channel model and ISI
- Optimal receiver design
- Nyquist criterion, Nyquist rate, and pulse shaping
- ML sequence estimation (optimal but high complexity)

Lecture 2: Linear Equalization (suboptimal but low(er) complexity)

Notes

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Geometric Model



- Received signal: $y(t) = \sum_n b[n]p(t - nT) + n(t)$
- Arbitrary receive filter $g_{RX}(t)$
 - suboptimal equalizers do not require an optimal front end
 - not matched to $p(t)$; often wideband filter, especially, if sampling is faster than the symbol rate ($m > 1$).
- Sampling time: $T_s = T/m$
 - $m = 1$: symbol spaced sampling
 - $m > 1$: fractionally spaced sampling
- Output of the sampler: $r[k] = (y \star g_{RX})(kT_s + \delta)$
- Sampled impulse response: $f[k] = (p \star g_{RX})(kT_s + \delta)$
- Noise at the output of the sampler: $w[k] = (n \star g_{RX})(kT_s + \delta)$
 - Colored noise with auto correlation/covariance

$$C_w(k) = 2\sigma^2 \int g_{RX}(t)g_{RX}^*(t - kT_s)dt$$

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Notes

Geometric Model

- A block of L received samples $\mathbf{r}[n]$ is used to decide one symbol $b[n]$.
- Model for the received vector

$$\mathbf{r}[n] = \mathbf{U}\mathbf{b}[n] + \mathbf{w}[n],$$

where

- $\mathbf{b}[n] = (b[n - k_1], \dots, b[n], \dots, b[n + k_2])^T$ is a length- K vector with $K = k_1 + k_2 + 1$; it includes all symbols $b[n']$ which contribute to $\mathbf{r}[n]$.
- $\mathbf{U} = [\mathbf{u}_{-k_1} \dots \mathbf{u}_{k_2}]$ is a $L \times K$ matrix with columns \mathbf{u}_i corresponding to the shifted impulse response $f[k]$.
- $\mathbf{w}[n]$ is zero-mean, proper complex Gaussian noise with covariance matrix \mathbf{C}_w

- Example²: $L = 3, k_1 = 1, k_2 = 0, K = 2$

$$\begin{array}{ccccccc} b[0]f[0] & & b[1]f[0] & & b[2]f[0] & & b[3]f[0] \\ & & + & & + & & + \\ & b[0]f[1] & & b[1]f[1] & & b[2]f[1] & \\ = & = & = & = & = & = & = \\ r[0] & r[1] & r[2] & r[3] & r[4] & r[5] & r[6] \end{array}$$

$= \mathbf{r}[2] \Rightarrow \mathbf{b}[2] = (b[1], b[2])^T$

$$\mathbf{r}[2] = \begin{bmatrix} r[3] \\ r[4] \\ r[5] \end{bmatrix} = \begin{bmatrix} f[1] & 0 \\ f[2] & f[0] \\ 0 & f[1] \end{bmatrix} \begin{bmatrix} b[1] \\ b[2] \end{bmatrix} + \begin{bmatrix} w[3] \\ w[4] \\ w[5] \end{bmatrix}$$

²Note that this choice of parameters may not be the optimal choice.

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Notes

Linear Equalization

- Alternative formulation of the received vector

$$\mathbf{r}[n] = b[n]\mathbf{u}_0 + \sum_{i=-k_1, i \neq 0}^{k_2} b[n+i]\mathbf{u}_i + \mathbf{w}[n]$$

- Linear equalization: correlate $\mathbf{r}[n]$ with a vector \mathbf{c} to produce a decision variable $Z[n] = \langle \mathbf{r}[n], \mathbf{c} \rangle$:

$$Z[n] = \mathbf{c}^H \mathbf{r}[n] = b[n]\mathbf{c}^H \mathbf{u}_0 + \underbrace{\sum_{i=-k_1, i \neq 0}^{k_2} b[n+i]\mathbf{c}^H \mathbf{u}_i}_{\text{residual ISI}} + \mathbf{c}^H \mathbf{w}[n]$$

- choose \mathbf{c} such that the signal component $b[n]\mathbf{c}^H \mathbf{u}_0$ is significantly larger than the residual ISI (if possible).
- Colored noise $\mathbf{c}^H \mathbf{w}[n]$ with covariance $\mathbf{c}^H \mathbf{C}_w \mathbf{c}$
- Hard decision or soft decision based on $Z[n]$

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Notes

Zero-Forcing Equalization

- The zero-forcing solution (if it exists) satisfies

$$\mathbf{c}^H \mathbf{u}_0 = 1 \quad \text{and} \quad \mathbf{c}^H \mathbf{u}_i = 0, \quad \text{for all } i \neq 0$$

or equivalently

$$\mathbf{c}^H \mathbf{U} = (0, \dots, 0, 1, 0, \dots, 0)^T = \mathbf{e}^T \quad \text{or} \quad \mathbf{U}^H \mathbf{c} = \mathbf{e}$$

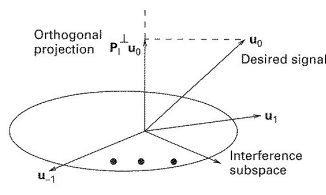
- The correlator \mathbf{c} should lie in the signal space spanned by the vectors $\{\mathbf{u}_i\}$
 - Components of \mathbf{c} outside the signal space contribute only noise.
 - \mathbf{c} should be a linear combination of the \mathbf{u}_i 's, i.e., $\mathbf{c} = \mathbf{U}\mathbf{a}$.
- It follows the zero-forcing equalizer as $\mathbf{c}_{ZF} = \mathbf{U}(\mathbf{U}^H \mathbf{U})^{-1} \mathbf{e}$

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Notes

Zero-Forcing Equalization

- Geometric interpretation of the signal space
 - Component in direction of the desired signal: \mathbf{u}_0
 - Interference subspace \mathbf{S}_I spanned by $\{\mathbf{u}_i, i \neq 0\}$



[U. Madhow, *Fundamentals of Dig. Comm.*, 2008]

- To suppress the interference completely, \mathbf{c} must be orthogonal to the interference subspace (only possible if $L > K - 1$): $\mathbf{c} = \alpha \mathbf{P}_I^\perp \mathbf{u}_0$
- With the normalization $\langle \mathbf{c}, \mathbf{u}_0 \rangle = 1$, we get the scale factor $\alpha = 1 / \|\mathbf{P}_I^\perp \mathbf{u}_0\|^2$.
- Noise variance σ_F^2 of the output noise:

$$v_{ZF}^2 = \sigma^2 \|\mathbf{c}\|^2 = \frac{\sigma^2}{\|\mathbf{P}_\perp^\top \mathbf{u}_0\|^2}$$

→ Noise enhancement with increasing α

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Notes

[illegible]

MMSE Equalization

- Minimization of the minimum mean squared error (MMSE)

$$\mathbf{c}_{MMSE} = \arg \min_{\mathbf{c}} \mathbb{E} \left[\underbrace{|\mathbf{c}^H \mathbf{r}[n] - b[n]|^2}_{\text{mean squared error (MSE)}} \right]$$

→ tradeoff between ISI suppression and noise enhancement

- Solution:

$$\mathbf{c}_{MMSE} = \mathbf{R}^{-1} \mathbf{p}$$

with

- autocorrelation matrix of the received signal $\mathbf{R} = E[\mathbf{r}[n]\mathbf{r}^H[n]]$
- cross-correlation vector $\mathbf{p} = E[b^*[n]\mathbf{r}[n]]$

- Proof: by solving $\nabla_{\mathbf{c}^H} \mathbb{E}[|\mathbf{c}^H \mathbf{r}[n] - b[n]|^2] = 0$
- Useful performance measure for linear equalizers: signal-to-interference ratio (SIR)

$$\text{SIR} = \frac{\sigma_b^2 |\langle \mathbf{c}, \mathbf{u}_0 \rangle|^2}{\sigma_b^2 \sum_{i \neq 0} |\langle \mathbf{c}, \mathbf{u}_i \rangle|^2 + \mathbf{c}^H \mathbf{C}_w \mathbf{c}}$$

- Important properties:
 - The MMSE equalizer maximizes the SIR.
 - At high SNR, the MMSE equalizer specializes to the ZF equalizer.

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Notes

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Adaptive Implementations

- Channel knowledge required for equalization
 - ZF: $\mathbf{U} \Rightarrow$ channel impulse response required
 - MMSE: $\mathbf{R}, \mathbf{p} \Rightarrow$ auto and cross correlation of the received vector and the data
- Training and decision directed modes
 - training phase: training sequences (known to the receiver) are transmitted; the receiver estimates parameters of the EQ.
 - data phase: data are transmitted, and the receiver uses the EQ based on the estimated parameters.

Least squares algorithm

- MMSE with empirical averages of \mathbf{R} and \mathbf{p} (N training sequences)

→ $\mathbf{c}_{LS} = \hat{\mathbf{R}}^{-1} \hat{\mathbf{p}}$ with

$$\hat{\mathbf{R}} = \frac{1}{N} \sum_{n=1}^N \mathbf{r}[n] \mathbf{r}^H[n] \quad \text{and} \quad \hat{\mathbf{p}} = \frac{1}{N} \sum_{n=1}^N b^*[n] \mathbf{r}[n]$$

Least mean squares algorithm

- Gradient descent: $\mathbf{c}[k] = \mathbf{c}[k-1] - \underbrace{\mu \mathbb{E}[\mathbf{r}[n](\mathbf{r}^H[n]\mathbf{c}[k-1] - b^*[n])]}_{\nabla_{\mathbf{c}^*} J(\mathbf{c}[k-1])}$
- Replacing the expectation by the instantaneous value yields

$$\begin{aligned} \mathbf{c}[k] &= \mathbf{c}[k-1] - \underbrace{\mu \mathbf{r}[n](\mathbf{r}^H[n]\mathbf{c}[k-1] - b^*[n])}_{e^*[k]} \\ &= \mathbf{c}[k-1] + \mu e^*[k] \mathbf{r}[n] \end{aligned}$$

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Notes

Performance Analysis

- The equalizer output can be written as

$$Z[n] = A_0 b[n] + \sum_{i \neq 0} A_i b[n+i] + W[n] \quad \text{with}$$

- amplitudes $A_i = \langle \mathbf{c}, \mathbf{u}_i \rangle$
- zero-mean Gaussian noise with variance $\nu^2 = \sigma^2 \|\mathbf{c}\|^2$
- BPSK system with hard decision based on $Z[n]$: $\hat{b}[n] = \text{sign}(Z[n])$
- Error probability

$$P_e = \Pr[\hat{b}[n] \neq b[n]] = \Pr[Z[n] > 0 | b[n] = -1]$$

due to symmetry.

- Conditional error probability for given ISI bits $\mathbf{b}_I = \{b_{n+i}, i \neq 0\}$

$$P_{e|\mathbf{b}_I} = \Pr[Z[n] > 0 | b[n] = -1, \mathbf{b}_I]$$

$$= \Pr[W[n] > A_0 - \sum_{i \neq 0} A_i b[n+i]] = Q\left(\frac{A_0 - \sum_{i \neq 0} A_i b[n+i]}{\nu}\right)$$

- Error probability averaged over ISI: $P_e = \mathbb{E}_{\mathbf{b}_I} \{P_{e|\mathbf{b}_I}\}$
- Alternative approximation: model ISI as Gaussian random variable with variance $\nu_I^2 = \sum_{i \neq 0} A_i^2$

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Notes
