

Equalization

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The Band-limited Baseband Transmission

Efficient Transmission Techniques

- Discrete PAM (Pulse Amplitude modulation)
 - ▶ to use a discrete set of possible amplitude level
- Linear Modulation
 - ▶ to conserve the bandwidth

The Band-limited Baseband Reception

- 1 demodulated
 - 2 synchronously sampled
 - 3 decision
- when high SNR
 - ▶ the number of detectable amplitude level (a discrete set)
 - ▶ depend on isi rather than channel noise
 - for a known channel characteristics
 - ▶ always possible to minimize isi
 - ▶ by using a proper pair of transmit and receive filters
 - ▶ to control the pulse shape

Why Equalization

In practice,

- no prior knowledge of the exact channel characteristics
- the imperfection of the implemented pulse shaping filters
- there will be always be some residual distortion
 - ▶ the isi is the limiting factor on the max possible data rate
- To cope with the intrinsic residual distortion
 - ▶ euqalization is used

The Traversal Filter

Two Linear Functions

- 1 pulse shaping
- 2 equalization of residual distortion

The Traversal Filter

- the combined and adjustable filter structure
- Delay Line - T second apart (symbol duration)
- Adjustable Weights - connected to the delay taps
- Summer - adding successively delayed and weighted input signal

Zero-Forcing Equalization

Two Subsystems

- 1 the impulse response $c(t)$:
combined transmit filter and communication channel
- 2 the impulse response $h_{eq}(t)$:
pulse shaping and residual distortion equalization

The Trasversal Fitler

- the structural symmetry - $(2N + 1)$ points
- $w_{-N}, \dots, w_{-1}, w_0, w_1, \dots, w_N$: filter coefficients

Discrete Convolution Sum

- $$h_{eq}(t) = \sum_{k=-N}^{+N} w_k \delta(t - kT)$$
- $$p(t) = c(t) \star h_{eq}(t) = c(t) \star \sum_{k=-N}^{+N} w_k \delta(t - kT)$$
- $$p(t) = \sum_{k=-N}^{+N} w_k c(t) \star \delta(t - kT) = \sum_{k=-N}^{+N} w_k c(t - kT)$$
- $$p(t = iT_b) = \sum_{k=-N}^{+N} w_k c((i - k)T)$$
- $$p_i = \sum_{k=-N}^{+N} w_k c_{i-k}$$

Simultaneous Equations

- $$h_{eq}(t) = \sum_{k=-N}^{+N} w_k \delta(t - kT)$$

Reference

[1] S. Haykin, M Moher, “Introduction to Analog and Digital Communications”, 2ed