



Helsinki University of Technology
Signal Processing Laboratory

S-38.411 Signal Processing in Telecommunications I Spring 2000

Lecture 8: Nonlinear receivers 1: DFE equalizers

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Timetable



- L1 Introduction; models for channels and communication systems
- L2 Channel capacity
- L3 Transmit and receive filters for bandlimited AWGN channels
- L4 Optimal linear equalizers for linear channels 1
- L5 Optimal linear equalizers for linear channels 2
- L6 Adaptive equalizers 1
- L7 Adaptive equalizers 2
- L8 Nonlinear receivers 1: DFE equalizers**
- L9 Nonlinear receivers 2: Viterbi algorithm
- L10 GL1: DSP for Fixed Networks / *Matti Lehtimäki, Nokia Networks*
- L11 GL2: DSP for Digital Subscriber Lines / *Janne Väänänen, Tellabs*
- L12 GL3: DSP for CDMA Mobile Systems / *Kari Kalliojärvi, NRC*
- L13 Course review, questions, feedback
- E 24.5. (Wed) 9-12 S4 **Exam**

Contents of Lecture 8



Nonlinear receivers 1: DFE equalizers

I. Basic idea of decision-feedback equalization

II. Design of DFE filters

III. Adaptive DFE

IV. Error propagation



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I. Basic idea of decision-feedback equalization

Basic idea of DFE



- ◆ In the previous lectures, *linear* equalizers and their adaptive implementations have been studied
- ◆ Linear MMSE equalizer in stochastic gradient (=LMS) adaptive implementation is simple, efficient and robust
- ◆ Problem: performance is not always good enough
 - noise enhancement in channels with zeroes
 - long impulse responses are a problem
- ◆ Goal of this lecture:
Improve the linear equalizer by simple nonlinear modifications

Basic idea of DFE...

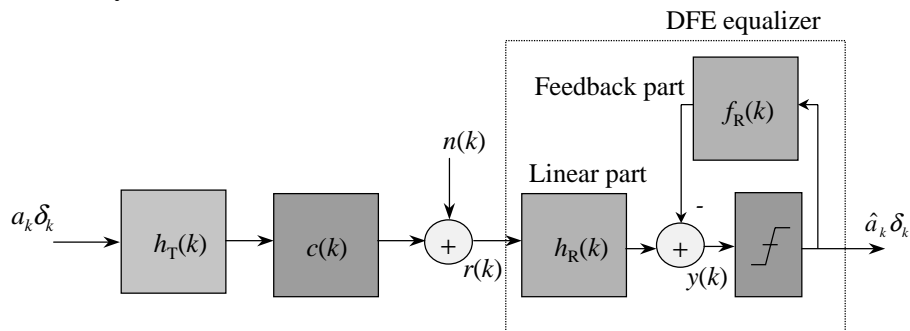


- ◆ Linear equalizer processes input signal samples $r(k)$ only
- ◆ Noise always limits the performance
- ◆ Noise enhancement problem (particularly with ZF equalizer)
- ◆ Basic problem in linear filtering: desired signal and noise processed together
- ◆ New approach:
Utilize previous symbol decisions \hat{a}_k to cancel ISI!

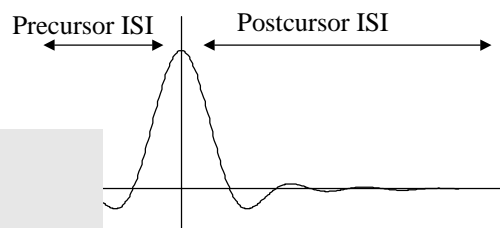
Basic idea of DFE...



- ◆ Basic structure of a DFE equalizer
(symbol-rate discrete-time model):



Basic idea of DFE...



- ◆ Channel impulse response:
 - at any decision instant, there is ISI contribution from some ‘future’ symbols (precursor ISI) and past symbols (postcursor ISI)
 - in causal systems, postcursor ISI can be moved by subtracting the weighted symbol value (*decision feedback*)



II. Design of DFE filters

Design of DFE filters...

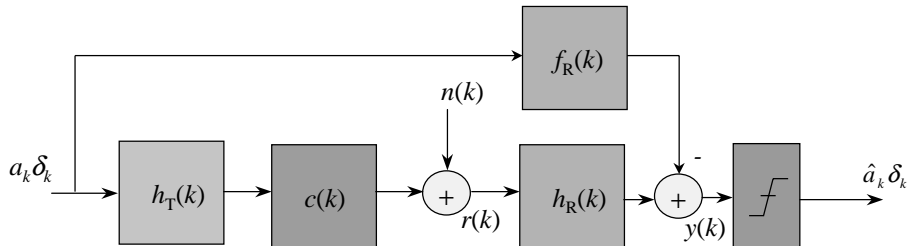


- ◆ Both $h_R(k)$ and $f_R(k)$ are *linear* filters
- ◆ Complete filtering system is *nonlinear*, because nonlinear operation (symbol decision) is in the feedback loop
- ◆ For the design of DF Equalizer, both linear part (feed-forward) filter and feedback filter need to be determined
- ◆ For easier analysis and design, the system is *linearized* by assuming all decisions correct ($\hat{a}_k = a_k$)

Design of DFE filters...



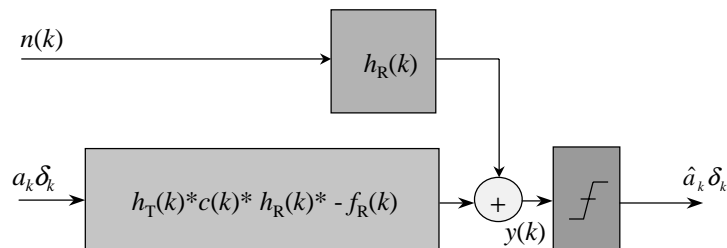
- ◆ Equivalent block diagram for *linearized* system:



Design of DFE filters...



- ◆ Simplified version, with separated signal and noise paths:



Design of DFE filters...



- ◆ Zero-forcing DFE:

$$h_T(k) * c(k) * h_R(k) - f_R(k) = \delta_k$$

$$\Leftrightarrow H_T(z)C(z)H_R(z) - F_R(z) = 1$$

- ◆ Several possible solutions for linear $h_R(k)$ and feedback filters $f_R(k)$ in the general case (infinite-length filters)
- ◆ Noise gain depends on choice of linear filter $h_R(k)$

Design of DFE filters...



- ◆ MMSE-DFE: minimize total mean square error $E[e^2(k)]$

$$\begin{aligned} e(k) &= y(k) - a_k \\ &= [h_T(k) * c(k) * h_R(k) - f_R(k) - \delta_k] * a_k + h_R(k) * n(k) \end{aligned}$$

- ◆ Assume independent signal and noise:

$$\begin{aligned} E[e^2(k)] &= E\left[\left([h_T(k) * c(k) * h_R(k) - f_R(k) - \delta_k] * a_k\right)^2\right] \\ &\quad + E\left[\left(h_R(k) * n(k)\right)^2\right] \end{aligned}$$

Design of DFE filters...



- ◆ Express MSE in frequency domain:

$$E[e^2(k)] = \int |H_T(f)C(f)H_R(f) - F_R(f) - 1|^2 S_x(f) df \\ + \int |H_R(f)|^2 S_n(f) df$$

- ◆ MSE = MIN when:

$$\int |H_R(f)|^2 S_n(f) df = \text{MIN}$$

$$\text{with } H_T(f)C(f)H_R(f) - F_R(f) = 1$$

Design of DFE filters...



Procedure for MMSE-DFE filter design:

- 1) Design linear filter $H_R(f)$ so that noise is minimized
- 2) Design feedback $F_R(f)$ filter so that ISI = zero

- ◆ In the ideal case (infinite-length feedback filter), all ISI can be completely eliminated!
- ◆ In practice, only *postcursor* ISI from a finite number of previous decisions can be eliminated
- ◆ Precursor ISI can be reduced by linear (precursor) filter and adding delay in the system



III. Adaptive DFE filters

Adaptive DFE filters

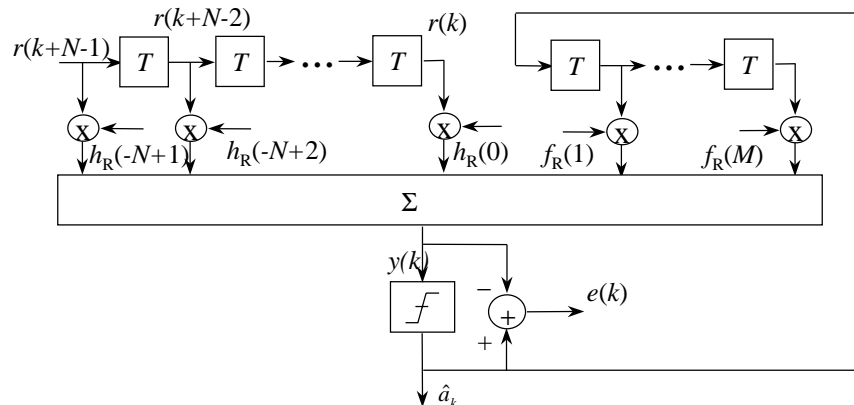


- ◆ Practical DFE filters ($h_R(k)$ and $f_R(k)$) are FIR filters
- ◆ Both linear and feedback filters are adjusted adaptively
- ◆ The adaptation can be done jointly for both $h_R(k)$ and $f_R(k)$ as if for a single FIR filter

Adaptive DFE filters...



- ◆ Signal flow diagram of adaptive DFE filter with $(N+M)$ taps:



Adaptive DFE filters...



- ◆ Adaptive DFE algorithm:

Define *augmented* signal and coefficient vectors

$$\mathbf{h}_R^+ = [h_R(-N+1) \ \cdots \ h_R(0) \ -f_R(1) \ \cdots \ -f_R(M)]^T$$

$$\mathbf{r}^+(k) = [r(k+N-1) \ \cdots \ r(k) \ a_{k-1} \ \cdots \ a_{k-M}]^T$$

- ◆ DFE output signal: $y(k) = \mathbf{h}_R^{+T}(k) \mathbf{r}^+(k)$
- ◆ Error signal: $e(k) = a_k - y(k) = a_k - \mathbf{h}_R^{+T}(k) \mathbf{r}^+(k)$

Adaptive DFE filters...



- ◆ Stochastic gradient algorithm for DFE:

$$\begin{aligned}\mathbf{h}_R^+(k+1) &= \mathbf{h}_R^+(k) - \frac{\beta}{2} \nabla_{\mathbf{h}_R} e^2(k) \\ &= \mathbf{h}_R^+(k) + \beta e(k) \mathbf{r}^+(k)\end{aligned}$$

- ◆ Formally and computationally simple like linear SG algorithm!
- ◆ Convergence properties similar to linear SG algorithm

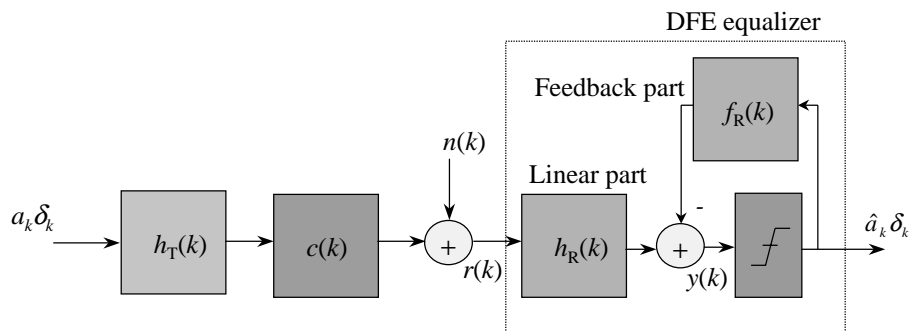


IV. Error propagation

Error propagation



- ◆ Linearized DFE model assumes all decisions *correct*
- ◆ What happens if they are not?



Error propagation...



- ◆ One decision error in DFE causes a *burst of new errors*
- ◆ The errors only stop after M (= order of feedback filter) *consecutive* correct decisions
- ◆ It can be shown (see Lee-Messerschmitt Appendix 10-A) that this happens after K symbols in the average, where

$$K = 2(2^M - 1)$$

- ◆ This gives average error probability

$$P_e = 2^M P_{e,0}$$

where $P_{e,0}$ is the error probability with no error propagation

Error propagation...



- ◆ Error propagation is the major problem in DFE
- ◆ It can be kept in control by keeping the error probability low (with other system choices) and keeping the feedback filter short enough

- ◆ Note! The error probability after DFE *cannot* be improved by error-correcting coding! (Why?)

- ◆ Alternative for DFE: *Tomlinson-Harashima precoding*
 - move feedback part in the transmitter
 - more in the DSL guest lecture!

Summary



Today we discussed:

Nonlinear receivers 1: DFE equalizers

I. Basic idea of decision-feedback equalization

II. Design of DFE filters

III. Adaptive DFE

IV. Error propagation

Next week: Viterbi algorithm