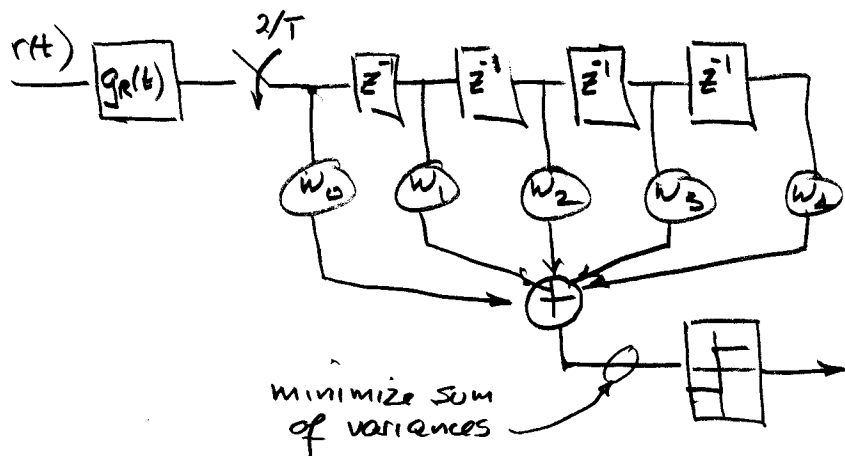


The Linear Equalizer - MMSE

- Insistence on zero interference is costly in terms of noise. Why not allow a little interference in exchange for a big reduction in noise?

In fact why not minimize the sum of the ISI and noise variances?



- Form a MMSE estimate of (say) a_0 as a LC of samples $y(-N\frac{T}{2}), y(-(N-1)\frac{T}{2}), \dots, y(-\frac{T}{2}), y(0), y(\frac{T}{2}), \dots, y(N\frac{T}{2})$
- \mathbf{y}^T

We know how to do this already!

$$\hat{a}_0 = \underline{w}^T \mathbf{y} \quad \underline{w}_{opt} = R_y^{-1} P_{ay}$$

$$R_y = E[\mathbf{y} \mathbf{y}^T] \quad P = E[a_0 \mathbf{y}]$$

- Look at the formation of the arrays. We have

$$y(t) = \sum_k a_k x(t-kT) + v(t), \quad y(mT/2) = \sum_k a_k x(mT/2 - kT) + v(mT/2)$$

Suppose span of impulse response is $2L+1$ samples

$$\underbrace{x(-LT/2), x(-(L-1)T/2), \dots, x(-T/2), x(0), \dots, x(LT/2)}_{x^T}$$

Then

$$\underline{y} = \begin{matrix} x^T & & 0 \\ \underline{x}^T & & 0 \\ 0 & x^T & 0 \\ 0 & & x^T & 0 \\ & & & x^T \\ & & & & x^T \\ & & & & & x^T \end{matrix} \underline{a} + \underline{v}$$

← x^T slipped by 1 sample, row to row

$X, (2N+1) \times (N)$

and $\underline{y} = X\underline{a} + \underline{v}$

so $R_y = \overline{\underline{y}\underline{y}^T} = X \underbrace{\overline{\underline{a}\underline{a}^T}}_{= \sigma_a^2 I \text{ if iid data}} X^T + \underline{v}\underline{v}^T = \sigma_a^2 X X^T + R_v$

and $\underline{p} = \overline{a_0 \underline{y}} = \sigma_a^2 \tilde{\underline{x}}$ (time reversed \underline{x} , zero padded if necessary, to make length $2N+1$)

since $\overline{a_0 y(mT/2)} = \sum_k \overline{a_0 a_k} x(mT/2 - kT) + \overline{a_0 v(mT/2)}$
 $= \sigma_a^2 x(mT/2)$

- performance is much better than ZF in low SNR.

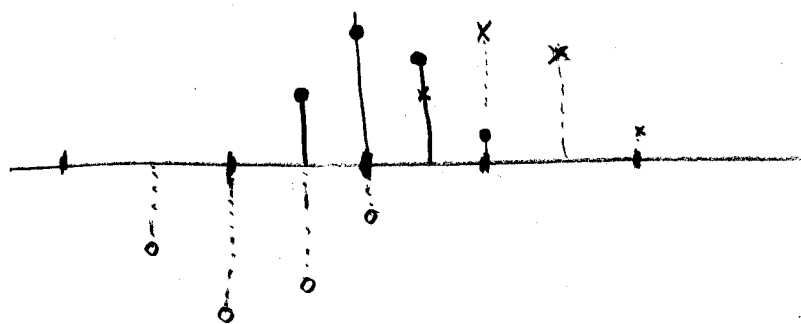
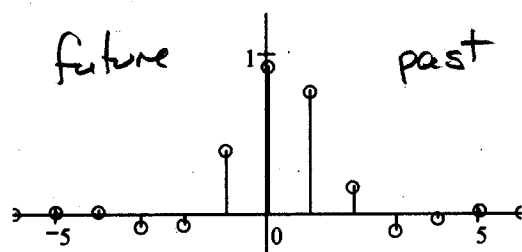
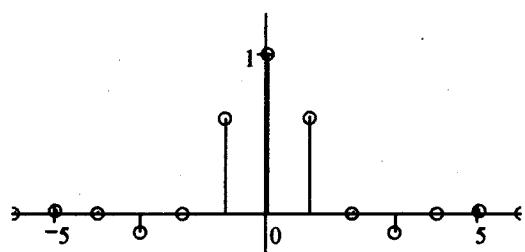
The Decision Feedback Equalizer

6.6.9

- This is the most widely used structure. It's easy and it works better than the linear equalizer.
- Recall that ISI contributions are typically from the past and future. This is root raised cos pulse and MF, with 1st order LPF channel.

$$f_c T = 100$$

$$f_c T = 1$$



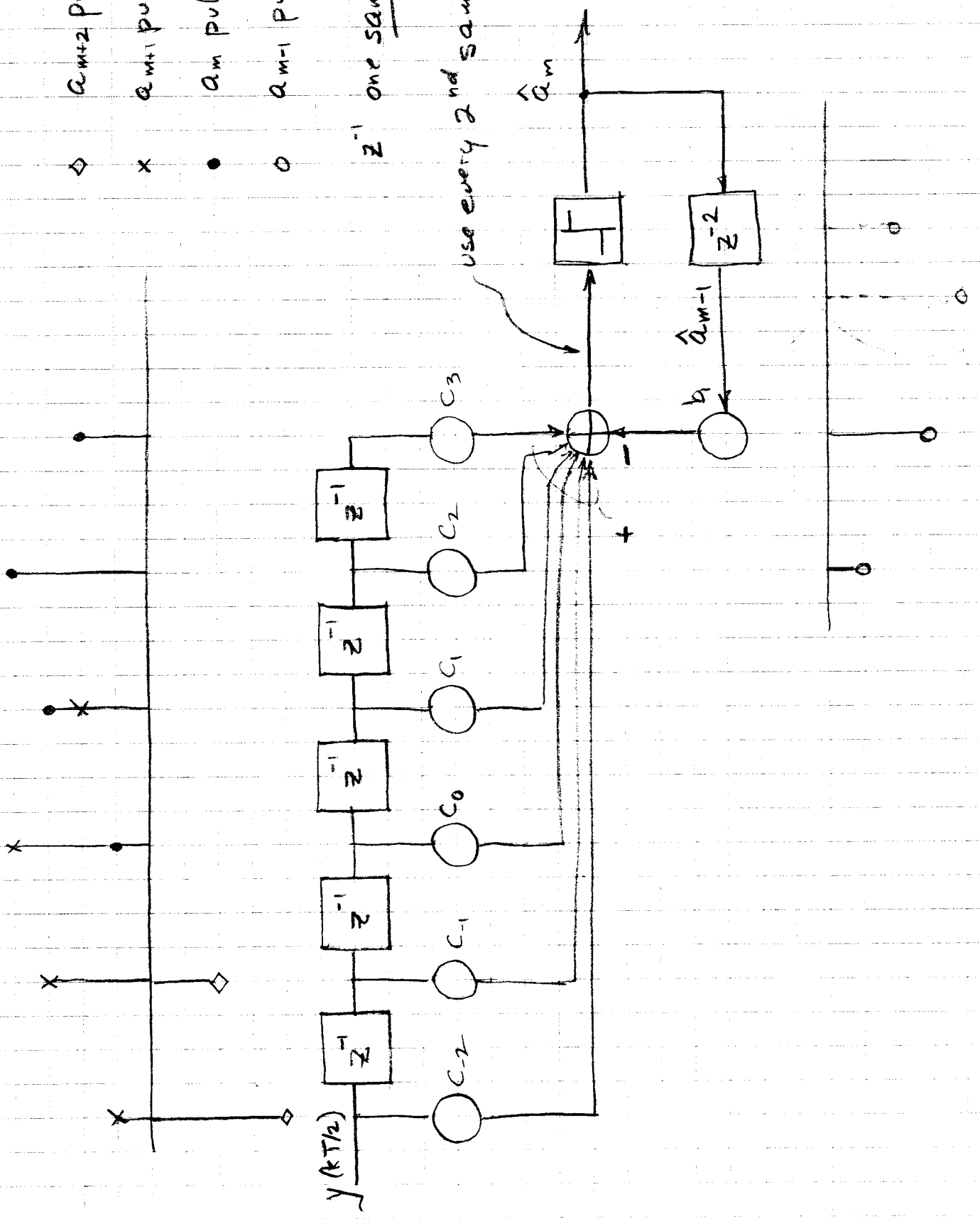
- x future pulse
- present pulse
- o past pulse

- If we happen to know the values of all neighbouring bits a_k , $k \neq m$, then we could subtract their contributions, leaving only $a_m x_{i-m}$ pulses and we could make a decision with only noise to cause errors.

- ◇ a_{m+2} pulse (future)
- × a_{m+1} pulse (future)
- a_m pulse
- a_{m-1} pulse (past)

z^{-1} one sample delay

use every 2nd sample



We do have past decisions. Proceed as if they are correct, and cancel their ISI,

- Each decision has its own weight to cancel the tail of its contribution
- The feedback section is symbol spaced
- The forward section is FT and coefficients selected by ZF or MMSE.

Channels A, B defined in text. A has roughly 10dB amplitude variation B has deep null.

In DFE, does a fed back error make another one more likely, leading to collapse?

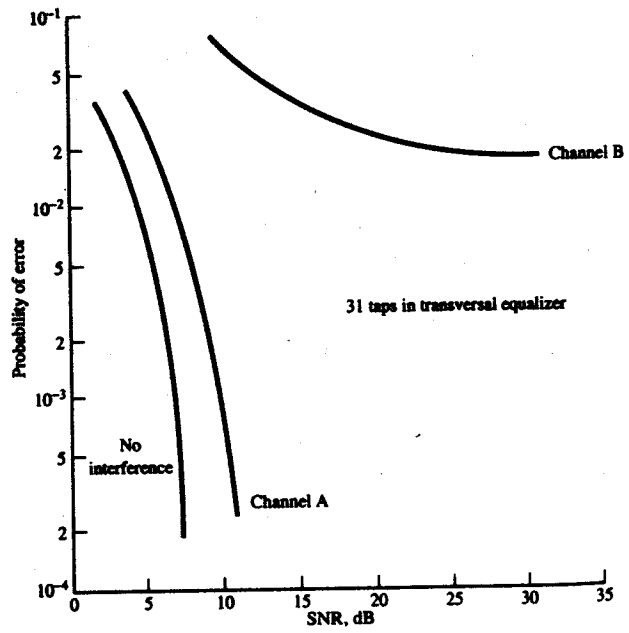


FIGURE 8.28. Error rate performance of linear MSE equalizer.

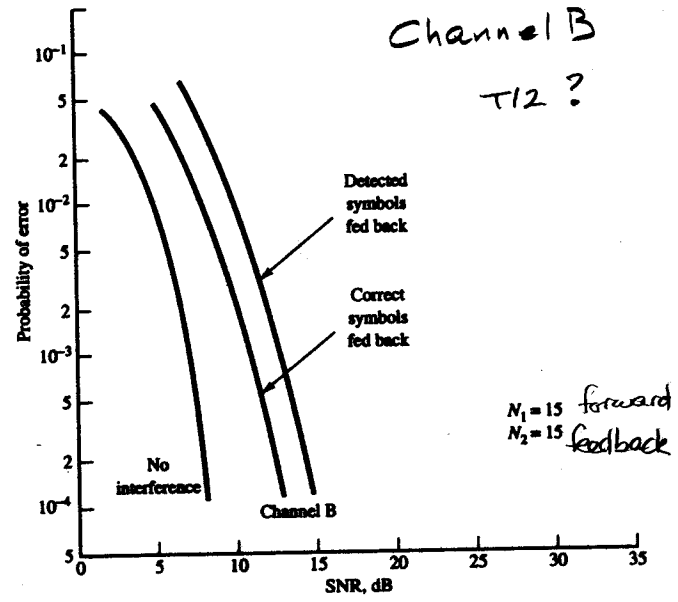


FIGURE 8.31. Performance of DFE with and without error propagation.

A DFE is particularly effective when there are deep nulls.