



STANFORD

Lecture 17
Precoders and Diversity
March 7, 2024

JOHN M. CIOFFI

Hitachi Professor Emeritus (recalled) of Engineering

Instructor EE379A – Winter 2024

Announcements & Agenda

Announcements

- PS8 is due March 12
 - Solutions immediately available, thus no late homework
 - Final distributed March 14, end of lecture. Due 5 pm on Friday 3/15.

Today

- Announcing our winner on the “chirp” sequence
- Transmit Precoders
- Partial-response channels
- Diversity and Turbo Equalizers



(Normalized) DFT

$$X_n = \frac{1}{\sqrt{N}} \cdot \sum_{k=0}^{N-1} x_k \cdot e^{-j\frac{2\pi}{N}kn}$$

$$x_k = \frac{1}{\sqrt{N}} \cdot \sum_{n=0}^{N-1} X_n \cdot e^{j\frac{2\pi}{N}kn}$$



Norm Tap Error

- Parsevals $\|\mathbf{h}\|^2 = \|\mathbf{H}\|^2$
 - And all other frequency-domain/time-domain vectors.

$$\begin{aligned} \text{NTE} &\triangleq E \left\{ \|\mathbf{h} - \hat{\mathbf{h}}\|^2 \right\} = E \left\{ \|\boldsymbol{\delta}\|^2 \right\} \\ &= E \left\{ \|\mathbf{H} - \hat{\mathbf{H}}\|^2 \right\} = E \left\{ \|\boldsymbol{\Delta}\|^2 \right\} \\ &= \sum_{n=0}^{\bar{N}-1} |H_n - \hat{H}_n|^2 = \sum_{k=0}^{\bar{N}-1} |h_k - \hat{h}_k|^2 \end{aligned}$$

$$SNR_{\hat{H},n} = \frac{R_{xx,n} \cdot |H_n|^2}{R_{ee,n} - R_{uu,n}} = \frac{SNR_n}{1/L} = L \cdot SNR \text{ (all } n\text{)}$$

- Same in all dimensions
 - The $L = 40$ leaves 0.1 dB gain-estimation error ($1+1/40 = 0.1$ dB)
- Constant magnitude in both domains
 - “White”
 - Lowest peak-to-average

$$\hat{H}_n = H_n + \sum_{l=1}^L \frac{U_{l,n}}{L \cdot X_{l,n}},$$

$$\Delta_n = - \sum_{l=1}^L \frac{U_{l,n}}{L \cdot X_{l,n}}.$$

$$\begin{aligned} E_n &= Y_n - \hat{H}_n \cdot X_n = \Delta_n \cdot X_n + U_n \\ &= U_n + \frac{1}{L} \cdot \sum_{l=1}^L U_{l,n} \cdot e^{j(\theta_n - \theta_{l,n})}. \end{aligned}$$

Good training sequence?

$$x_k = e^{j\frac{2\pi}{N} \cdot k^2}$$

(chirp)

DFT is

$$X_n = \frac{1}{\sqrt{N}} \cdot e^{-j\frac{\pi}{4N} \cdot n^2}$$

Another chirp



Noise Estimation

- Average the errors in frequency domain

$$\hat{\sigma}_n^2 = \frac{1}{L} \cdot \sum_{l=1}^L |E_{l,n}|^2$$

$$\text{var}(\hat{\sigma}_n^2) = \frac{1}{L^2} (3 \cdot L \cdot \sigma_n^4 - L \cdot (\sigma_n^2)^2) = \frac{2}{L} \sigma_n^4$$

- Noise miss only reduces with sqrt(L).

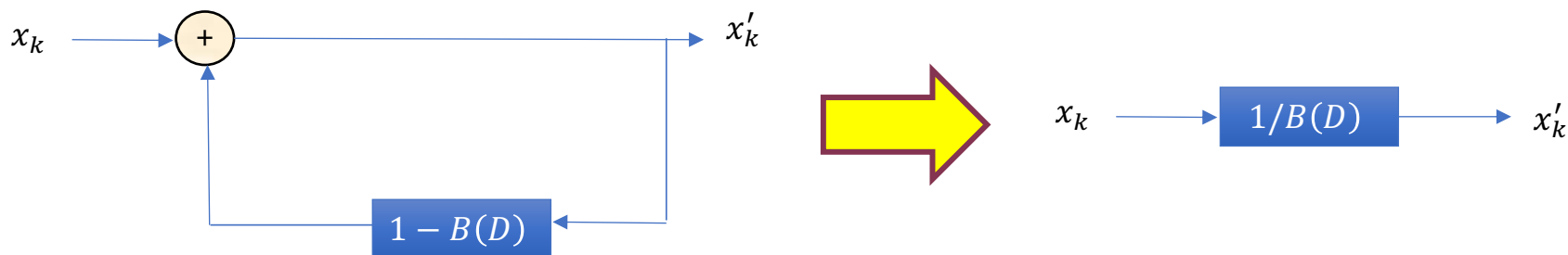
$$\sqrt{2/L} \cdot \sigma_n^2$$



Precoders

Section 3.8

Channel Inversion at the transmitter?

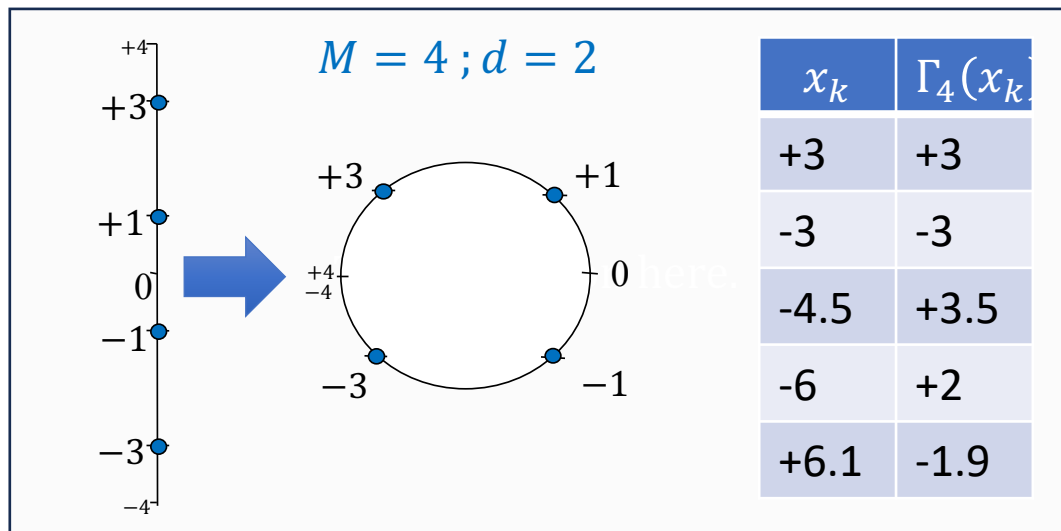


- So why not “pre-equalize” and remove ISI? Indeed, there is no noise enhancement at transmitter, so?
- Transmit energy exceeds the limit. Pre-inversion design reduces distance, and thus incurs a loss.
- But it still can be a good idea with a little help



Modulo Arithmetic

- Mod arith wraps the real line around circle of circumference $2M$; $\Gamma_M(x) = x - M \cdot d \cdot \left\lfloor \frac{x + \frac{M \cdot d}{2}}{M \cdot d} \right\rfloor$,



**Maps uniform
Distribution
[$-4i \ 4i$] $i \in \mathbb{Z}$
to another uniform
Distribution
[$-4 \ 4$]**

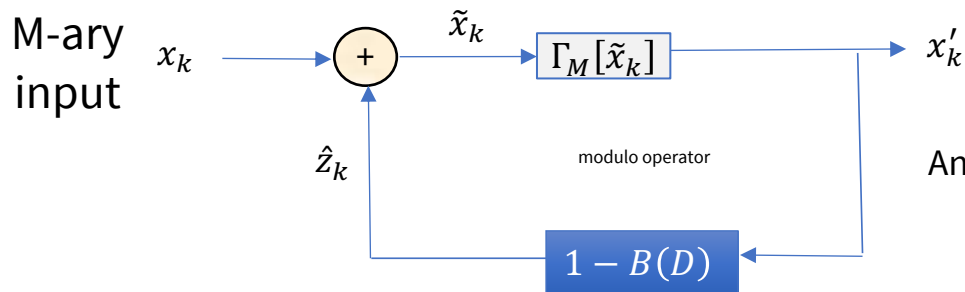
- Any two numbers with result on circle : $x \oplus_M y \triangleq \Gamma_M(x + y)$
 $x \ominus_M y \triangleq \Gamma_M(x - y)$

- Trivially: $\Gamma_M(x + y) = \Gamma_M(x) \oplus_M \Gamma_M(y)$
 $\Gamma_M(x - y) = \Gamma_M(x) \ominus_M \Gamma_M(y)$



The Precoder

- Tomlinson – Harashima (and probably others) illustrates basics, not used to instructor's knowledge.



Approximately uniform distribution over $[-\frac{Md}{2}, \frac{Md}{2})$
 And i.i.d. \sim same as quantization noise

$$\tilde{x}_k = x_k - \sum_{i=1}^{\infty} b_i \cdot x'_{k-i}$$

$$x'_k = \Gamma_M(\tilde{x}_k) = \Gamma_M\left(x_k - \sum_{i=1}^{\infty} b_i \cdot x'_{k-i}\right)$$

Implementation is for real baseband signals.

$$z_{U,k} = (x'_k + \sum_{i=1}^{\infty} g_{U,i} \cdot x'_{k-i}) + e_{U,k}$$

Another receiver modulo produces:

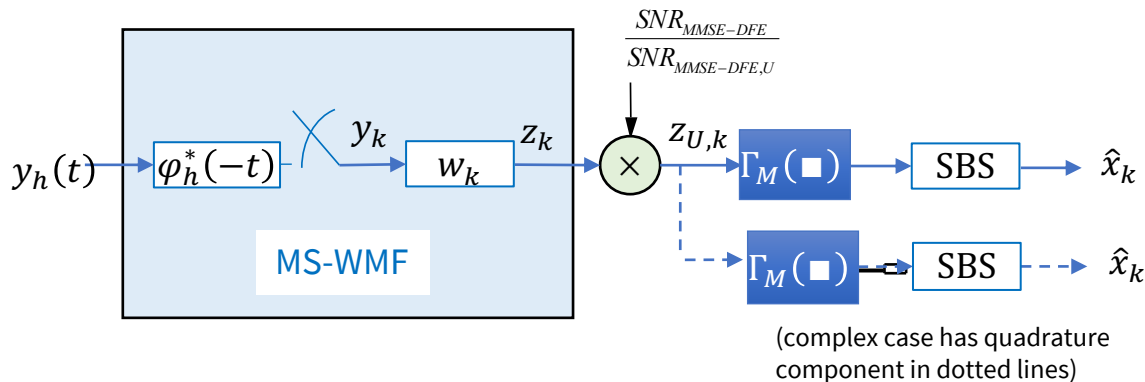
$$\Gamma_M(z_{U,k}) = x_k + e'_{U,k} \cong x_k + e_{U,k}$$

No errors at transmitter!

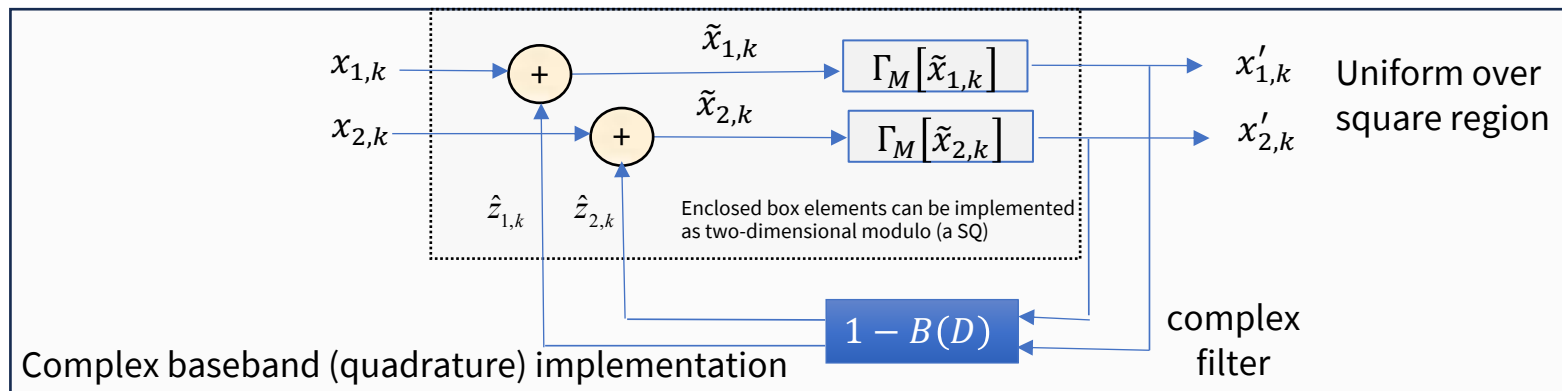
$$\begin{aligned} \Gamma_M[z_{U,k}] &= \Gamma_M\left[\Gamma_M\left(x_k - \sum_{i=1}^{\infty} g_{U,i} \cdot x'_{k-i}\right) + \sum_{i=1}^{\infty} g_{U,i} \cdot x'_{k-i} + e_{U,k}\right] \\ &= \Gamma_M\left[x_k - \sum_{i=1}^{\infty} g_{U,i} x'_{k-i} + \sum_{i=1}^{\infty} g_{U,i} \cdot x'_{k-i} + e_{U,k}\right] \\ &= \Gamma_M[x_k + e_{U,k}] \\ &= x_k \oplus_M \Gamma_M[e_{U,k}] \\ &= x_k + e'_{U,k} \end{aligned}$$



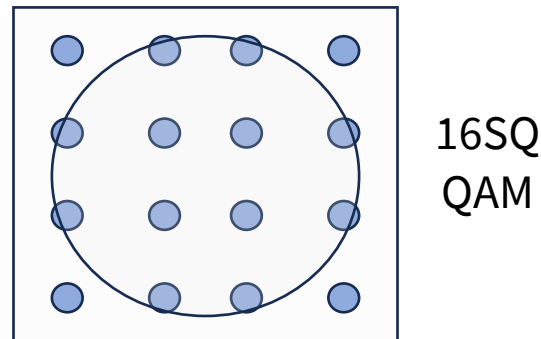
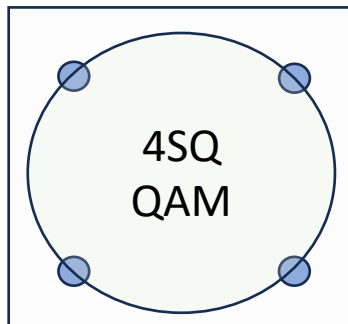
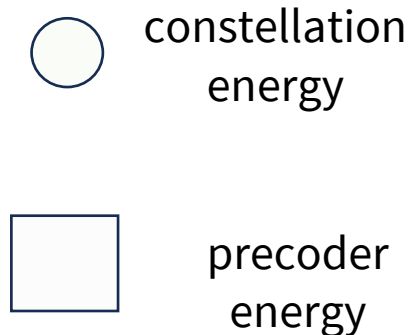
Receiver Block Diagram (no feedback)



- Receiver only needs (unbiased) feedforward filter (feedback already presubtracted in transmitter).



Transmit energy increases slightly

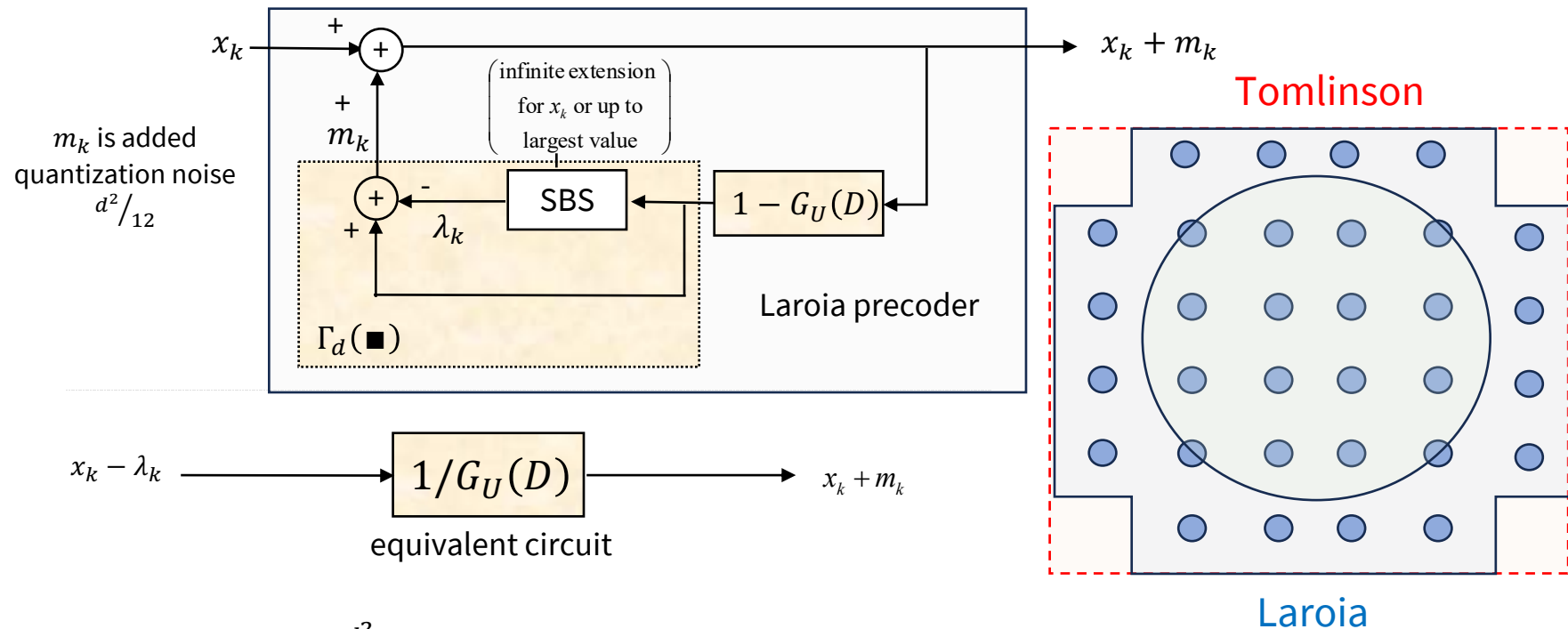


- The (Tomlinson) precoder output has (theoretically & practically) *continuous* uniform dist'n and
 - has uniform distribution over $\left[-\frac{M}{2}d, +\frac{M}{2}d\right]$. Mean-square (energy) $\frac{M^2 \cdot d^2}{12} >$ PAM's $\frac{(M^2-1) \cdot d^2}{12}$.
 - So there is a transmit energy effective loss of $\frac{M^2}{M^2-1}$.
 - For binary, this is 4/3 or 1.3 dB, as M increases, the loss becomes negligible.
- The same applies for SQ QAM with $M^2 \rightarrow M$. Shaping is not possible with **this** precoder.



Laroya Precoder – accommodates shaping

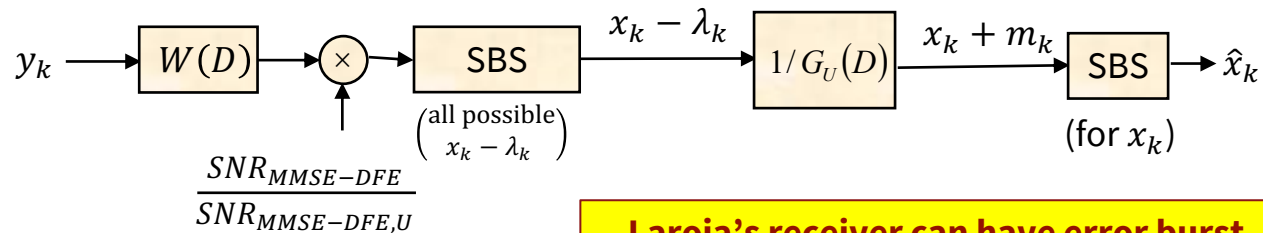
- **Laroya Precoder** inverts channel independent of constellation size (used in voiceband modem standards).



- Energy increase is $\frac{\bar{\epsilon}_x + \frac{d^2}{12}}{\bar{\epsilon}_x} = \frac{M^2 - 1}{M^2 - 2}$, worse than Tomlinson ($3/2$ for $M = 2$), preserves γ_s at large M .



Laroya's Receiver Circuit



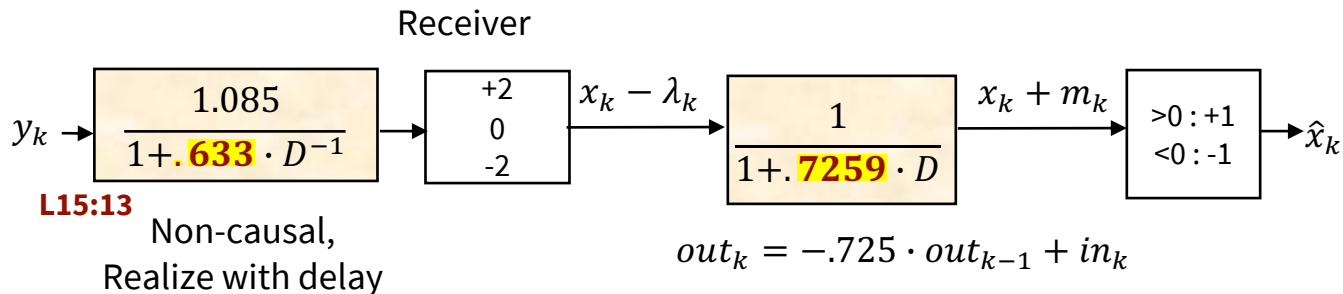
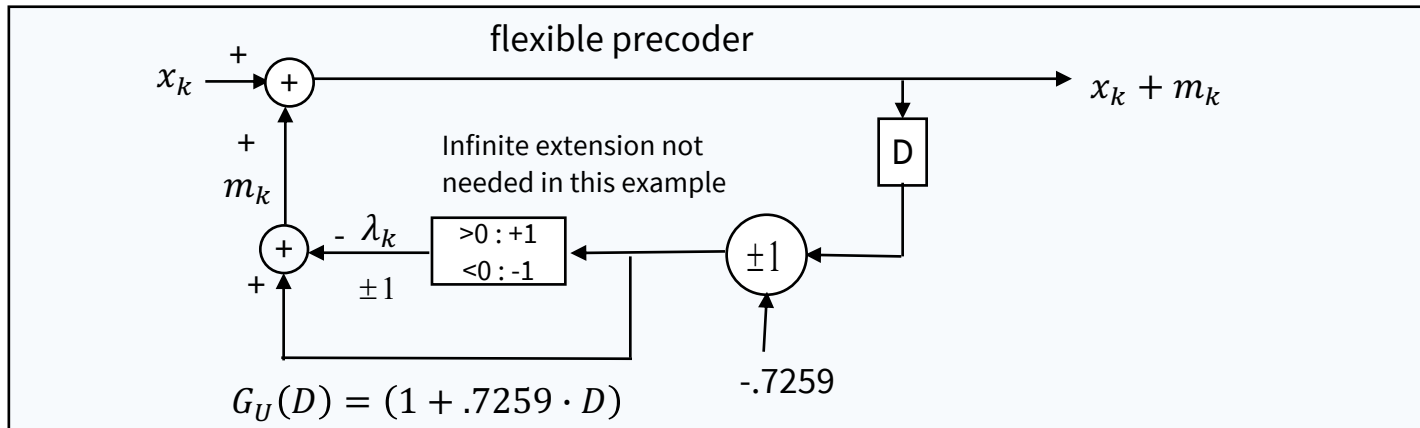
**Laroya's receiver can have error burst,
but no propagation beyond.
Model $G_u(D)$ as IIR, so then finite burst**

- LP was used in some wireline standards. Fortunately, there is a better solution – see EE379B (Chapter 4), which can also preserve γ_S .
- In MIMO systems a “GDFF” is used (also see 379B, Chapter 5), and its error propagation is very finite and limited, so those GDFF systems can be used largely without error-propagation concern.



Return to $1+.9D^{-1}$ example

- $N_f = \infty$; $\bar{\mathcal{E}}_x = 1$; $N_b = 1$; $\Delta = 1$; $\sigma^2 = .181$



L15:13

Non-causal,
Realize with delay

$$out_k = -.725 \cdot out_{k-1} + in_k$$

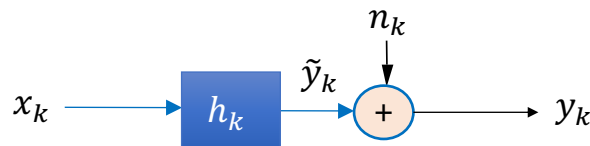
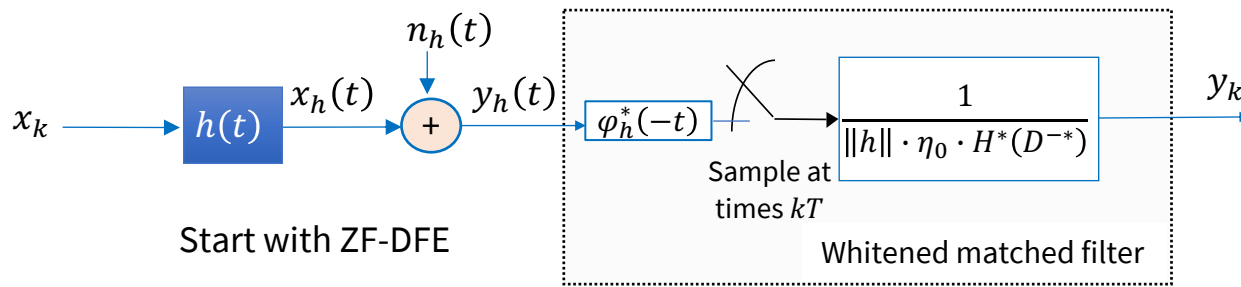
L17:13

Partial Response & Precoders

Section 3.8.2

Generally: “precoders” move decision feedback to the transmitter

Model (“equalize”) channel to $h_k \in \mathbb{Z}$



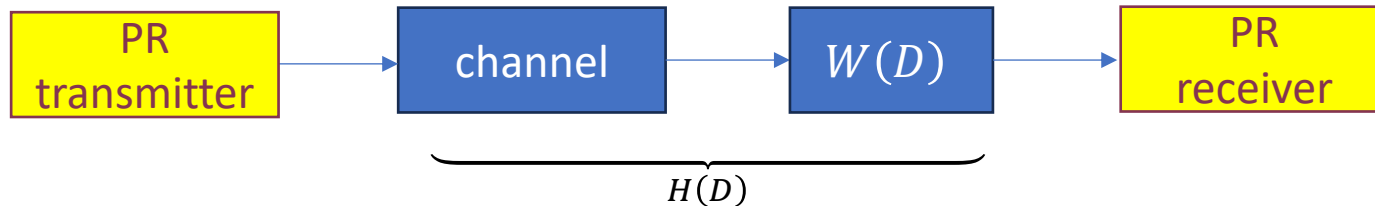
controlled ISI or **partial-response** channel, $h_{k<0} = 0, h_{k>v} = 0$ and $h_k \in \mathbb{Z}$

- The precoder simplifies, and there is no transmit energy loss if $h_k \in \mathbb{Z}$.
- The receiver has ZF-DFE performance for this integer channel with no error propagation.
- A Viterbi Detector for ISI can be used – increases performance (often, not always) to SNR_{MFB} .
 - Requires more receiver complexity.
 - Often imposed at transmitter, but there may be an equalizer(with loss) to force the integer channel coefficients.
 - Any external coding gain is largely lost unless iterative decoding is used (see Turbo Equalization at end of this L17.)



Partial-Response Channel Definition

- $H(D)$ has the **monic finite-length** form $H(D) = 1 + h_1 \cdot D + h_2 \cdot D^2 + \dots + h_\nu \cdot D^\nu$, **AND**
- $H(D)$ is **minimum-phase** (all roots/zeros outside or on unit circle), **AND**
- **Integer coefficients** $h_k \in \mathbb{Z}$.



- The channel may need equalization to such a response (See Section 3.13 – not taught).
 - If the $H(D)$ is pretty close to actual channel, this equalization loss can be small.
- We're going to assume here the blue part is already done.

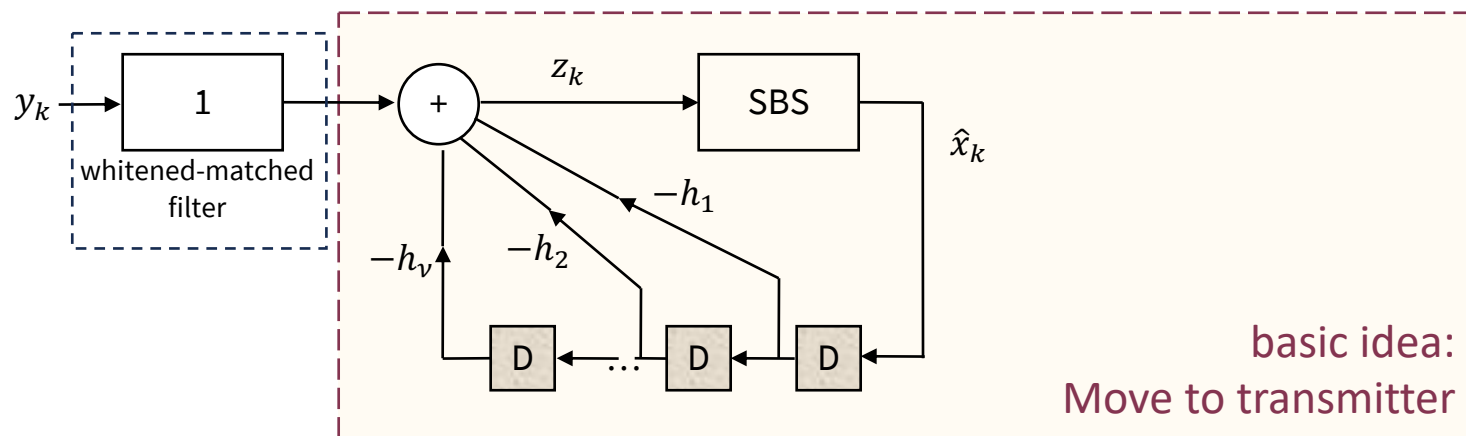


Precoding the PR's ZF-DFE is easy

Simple DFE operations

For $H(D) = 1 - D$, $z_k = y_k - \hat{x}_{k-1}$

For $H(D) = 1 + D - D^2 - D^3$, $z_k = y_k - \hat{x}_{k-1} + \hat{x}_{k-2} + \hat{x}_{k-3}$

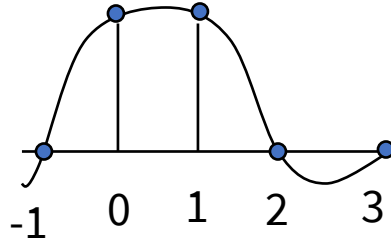
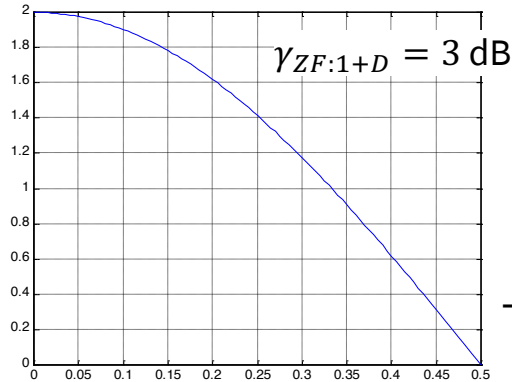


- The original channel is minimum phase, monic, and ZF-DFE becomes trivial.
- The ZF-DFE has error propagation.
- The precoder is easier, has no energy increase, and preserves the constellation.

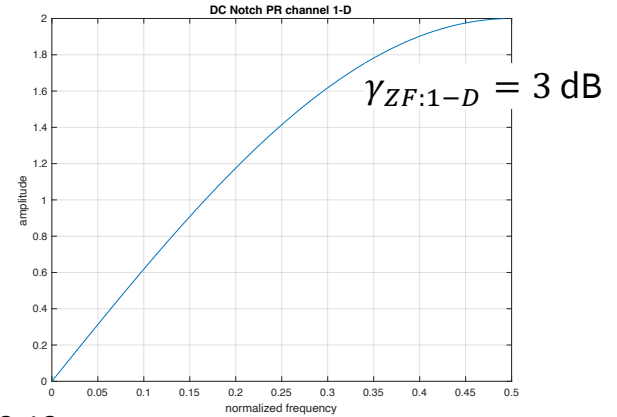


First, some popular PR channels

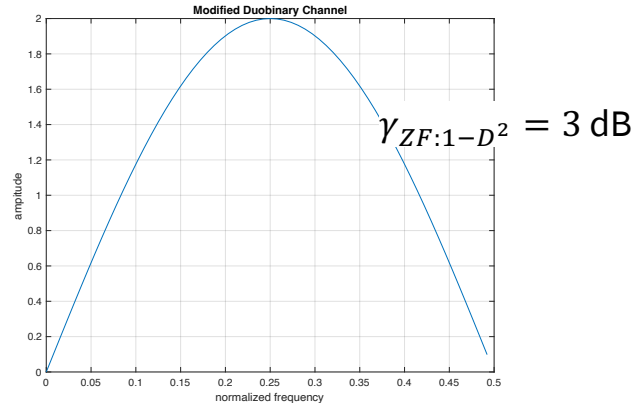
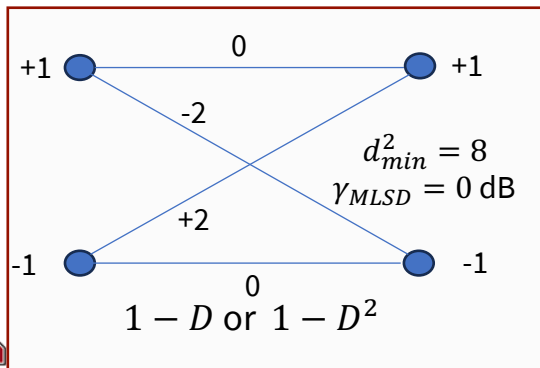
- Duobinary** $H(D) = 1 + D$ - models lowpass channel.



- DC-Notch** $H(D) = 1 - D$ - models highpass



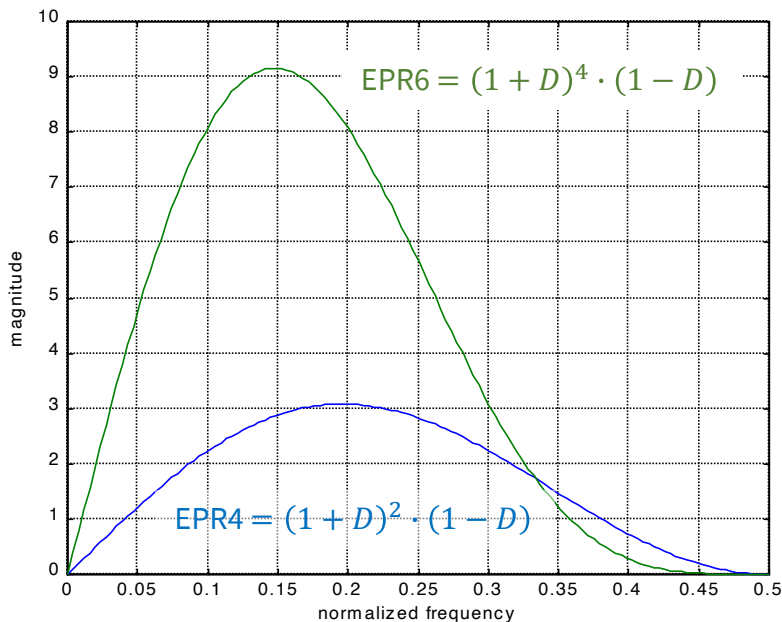
- Modified Duobinary** $H(D) = 1 - D^2$ - lowpass with DC notch "PR4/PRML" 😊, 3.13



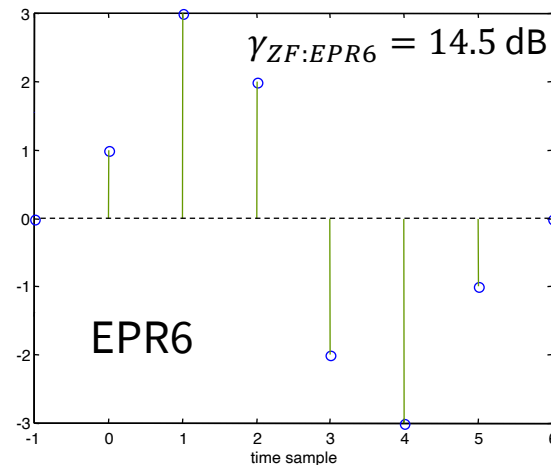
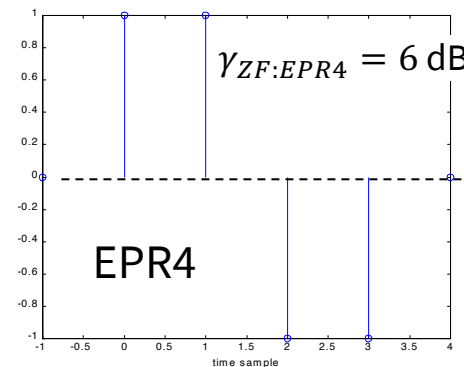
A potential project:
 run Chap 7 dmin
 program for MLSD
 with Euclidean distance
 replace bdistance.m.
 All have
 0 dB loss w.r.t.
BINARY MFB



Extended Partial Response (Thapar)



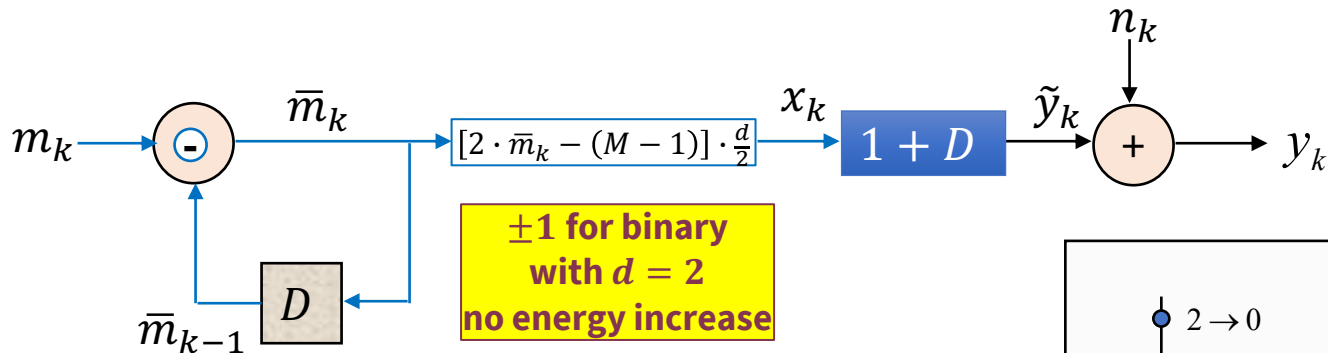
- $1+D$ factors increase ISI / lowpass effect.
- $H(D) = (1 + D)^n \cdot (1 - D)$.
- EPR is often used in “recording” channels (disk).



MLSD losses are
EPR4: 0 dB
EPR6: 2.2 dB
 ...
EPR8: 3.7 dB
EPR10: 4.5 dB



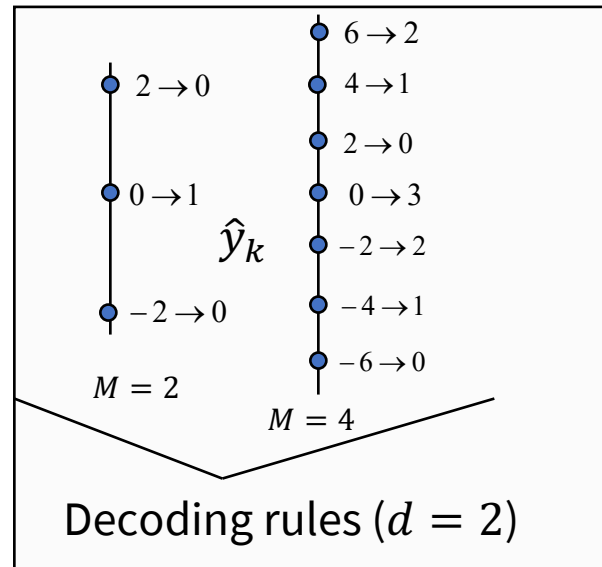
PR Precoders



- 1-D is almost the same with decoding rule flipped.
- General PR precoder is $\bar{m}_k = m_k \oplus_{i=1}^v (-h_i) \cdot \bar{m}_{k-i}$.
- General decode:

$$\hat{m}_k = \left(\frac{\hat{y}_k}{d} + \sum_{i=0}^v h_i \cdot \left\lfloor \frac{M-1}{2} \right\rfloor \right)_M$$

- General error prob: $P_e \leq 2 \cdot \left(1 - \frac{2}{M^{v+1}}\right) \cdot Q\left(\frac{d}{2\sigma}\right)$



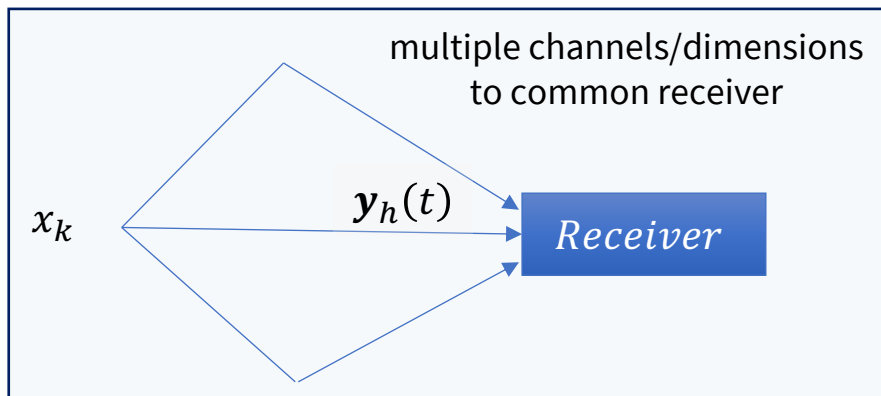
- For quadrature versions, see 3.8.6



Diversity Equalizers

Section 3.8.2

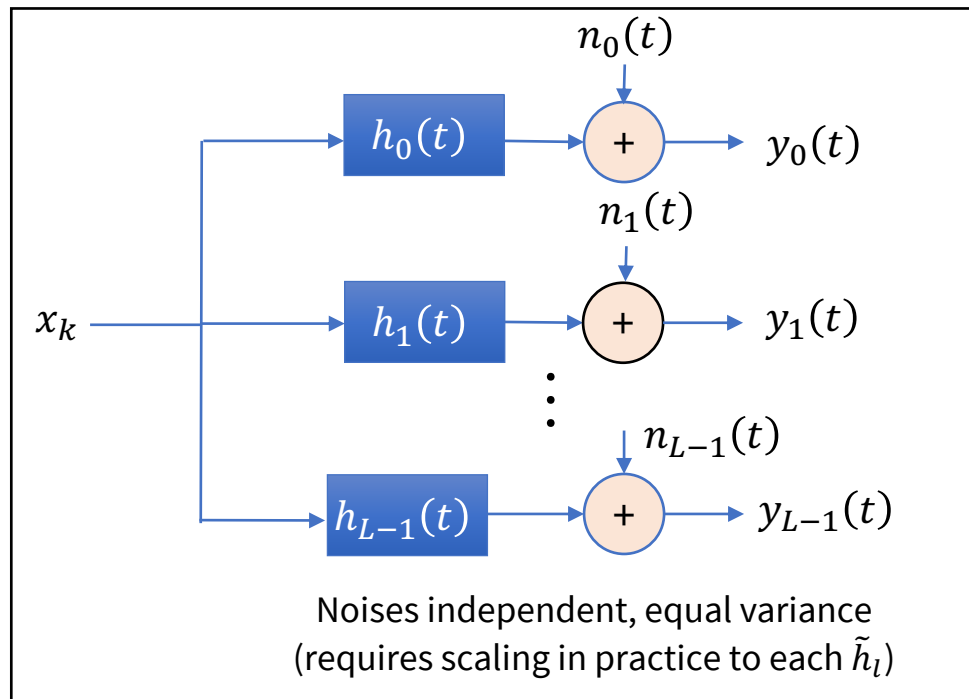
Multiple Received Signals



$$\mathbf{y}_h(t) = \sum_k x_k \cdot \mathbf{h}(t - kT) + \mathbf{n}_h(t)$$

- Multiple rcvr antennas (but single xmit antenna)
- Multiple repetitions (codes are form of diversity)
- MFB? (single no-ISI transmission)

$$\mathbf{y}_h(t) = x_0 \cdot \mathbf{h}(t) + \mathbf{n}_h(t)$$

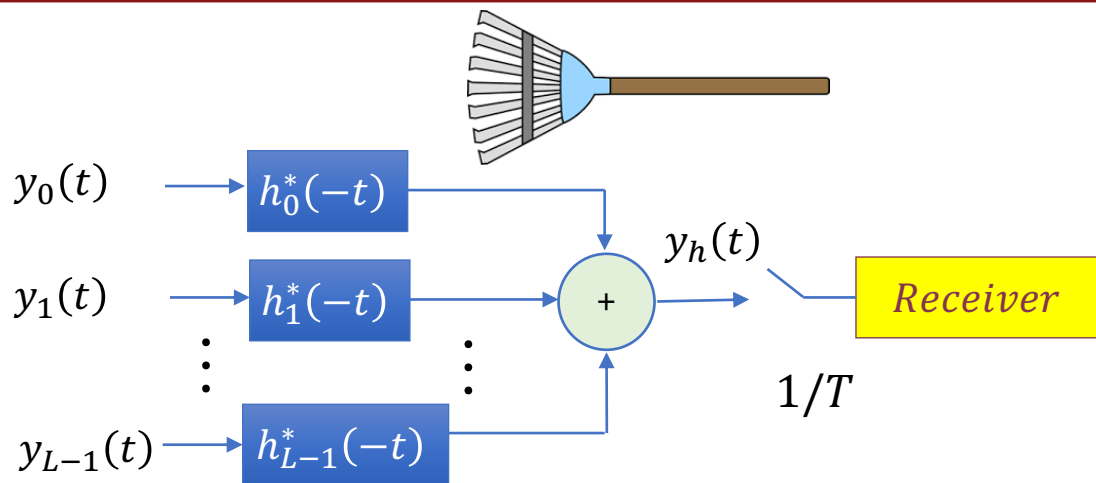


$$SNR_{MFB} = \frac{\bar{\mathcal{E}}_x \cdot \|\mathbf{h}\|^2}{\sigma^2}$$

$$\|\mathbf{h}\|^2 = \sum_{l=0}^{L-1} \|\tilde{h}_l\|^2$$



“RAKE” Receiver (P. Green)

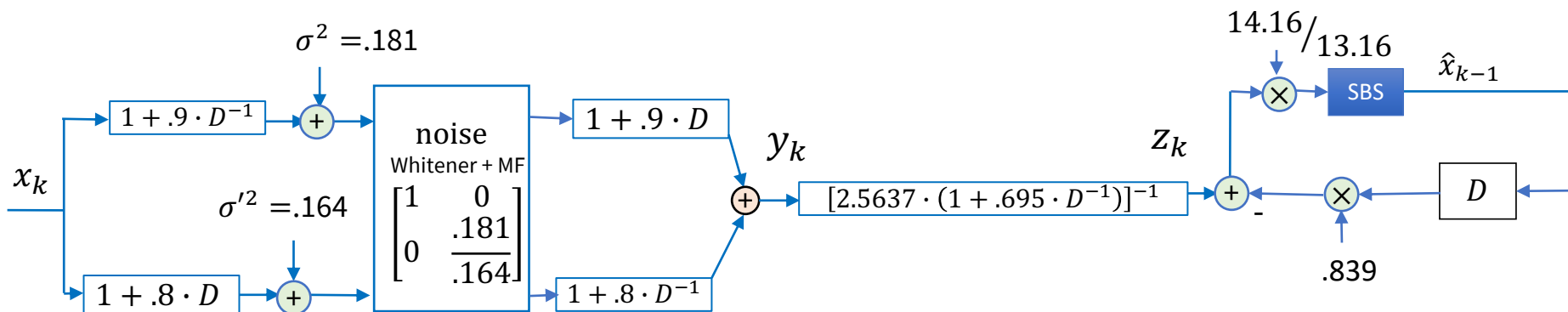


$$r(t) = \sum_{l=0}^{L-1} r_l(t) = \sum_{l=0}^{L-1} \tilde{h}_l(t) * \tilde{h}_l^*(-t) \quad r(t) = \|\mathbf{h}\|^2 \cdot q(t)$$

- Infinite-length theory applies to the scalar $Y(D) = X(D) \cdot \|\mathbf{h}\|^2 \cdot Q(D) + N(D)$.
- Receiver can be LE, DFE, MMSE or ZF ... (MLSD, etc).
- Can be implemented digitally with sufficiently high sampling rate preceding the matched filters.



2 Parallel Channels Example



$$\|\mathbf{h}\|^2 = 1.81 + \left(\frac{.181}{.164}\right) \cdot 1.64 = 2 \cdot 1.81 = 3.62$$

$$SNR_{MFB} = \frac{1 \cdot 3.62}{.181} = 20 \text{ (13) dB}$$

$$SNR_{MMSE-DFE,U} = .7082 \cdot 20 - 1 = 13.16 \text{ (11.15 dB)}$$

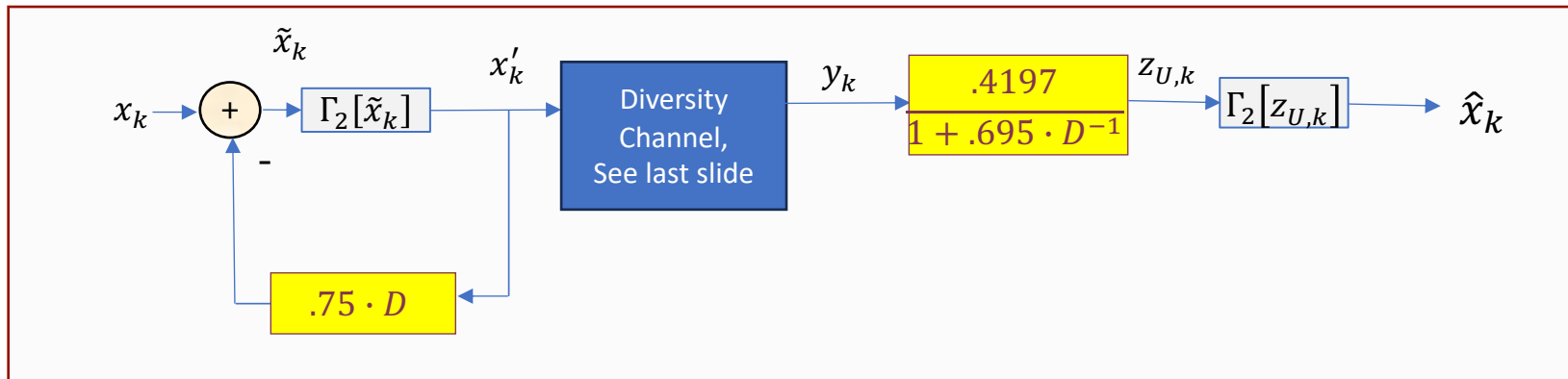
$$\begin{aligned} Q(D) &= \frac{1}{2(1.81)} \left[(1 + .9D^{-1})(1 + .9D) + \frac{.181}{.164}(1 + .8D)(1 + .8D^{-1}) \right] \\ &= .492D + 1 + .492D^{-1} \\ &= .589 \cdot (1 + .835D) \cdot (1 + .835D^{-1}) \\ \tilde{Q}(D) &= .492D + (1 + 1/20) + .492D^{-1} \\ &= .7082 \cdot (1 + .695D) \cdot (1 + .695D^{-1}) \end{aligned}$$

- Once equivalent sum-channel is found, subsequent receiver analysis is the same as earlier (LE, DFE, even MLSD)



Combine with Precoder?

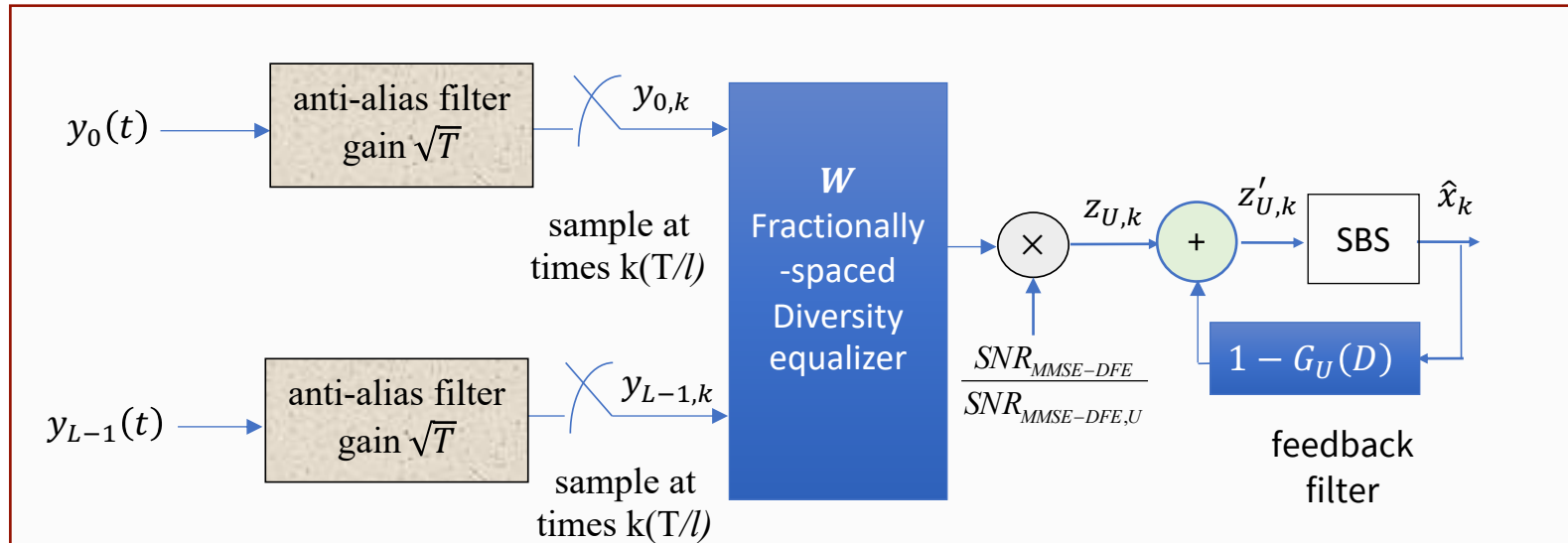
- Need $G_U(D) = 1 + \left(\frac{14.16}{13.16}\right) \cdot D = 1 + .75 \cdot D$; similarly $W_U(D) = \left(\frac{14.16}{13.16 \cdot 2.5637}\right) \cdot \frac{1}{1 + .695 \cdot D^{-1}} \cong \frac{.8}{1 + .7 \cdot D^{-1}}$



- Precoder loss (binary) is 4/3 or 1.3 dB, so SNR = 11.2 dB – 1.3 = 9.9 dB.
- Note that multiple paths, with appropriate receiver design, always improves the performance w.r.t. either path individually



Multidimensional FIR Equalizer/DFE



- The design can again follow the single-channel case theory
- or use DFERake.m .
- **“Diversity Receiver”** is the modern name for it.



DFE Rake Program

```
>> help dfeRAKE
function [dfseSNR,W,b]=dfeRAKE(l,h,nff,nbb,delay,Ex,noise);
DFE design program for RAKE receiver

Inputs
l   = oversampling factor
L is derived as No. of fingers in RAKE (number of rows in h)
h   = pulse response matrix, oversampled at l (size),
     each row corresponding to a diversity path
nff = number of feedforward taps for each RAKE finger
nbb = number of feedback taps
delay = delay of system <= nff+length of p - 2 - nbb
Ex   = average energy of signals
noise = noise autocorrelation vector (size L x l*nff)
NOTE: noise is assumed to be stationary, but may be spatially
correlated

outputs:
dfseSNR = equalizer SNR, unbiased in dB -----
```

```
>> hrake=[0.9000  1.0000   0
          0  1.0500  0.8400];
>> [snr,W,b] = dfeRAKE(1,hrake,6,1,5,1,[.181 zeros(1,5) ; .181 zeros(1,5)])

snr = 11.1465 dB
W =
    0.0213  -0.0439  0.0668  -0.0984  0.1430  0.3546
   -0.0027  0.0124  -0.0237  0.0382  0.4137  0.0000
b = 0.7022
```

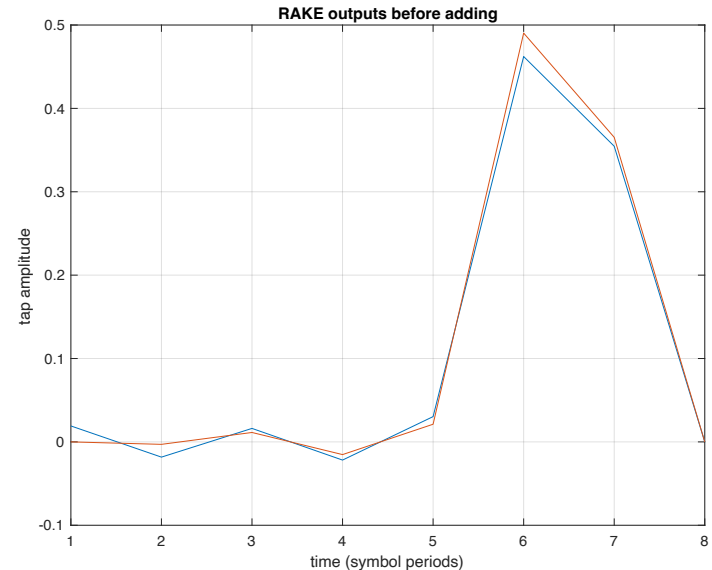
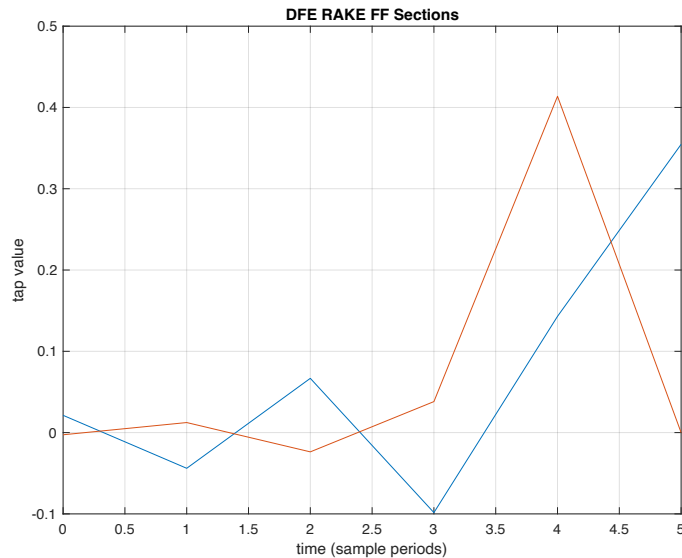
Student Project:
Add the -1 = delay
option to find
best delay.

- Few taps, matches infinite-length result.



DFE Rake Plots

- The MS-WMF's try to align to on another as well as in time to their respective paths.



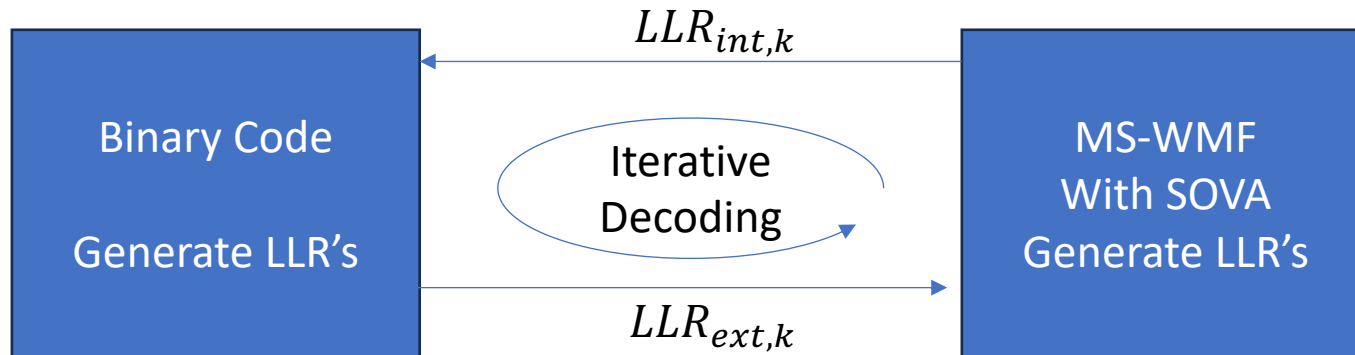
```
>> plot(conv(W(1,:),[.9 1 0]))  
>> hold  
Current plot held  
>> plot(conv(W(2,:),(1.81/1.64)*[0 1 .8]))
```

- The equalized channel clearly looks causal in last 3 positions, and the two outputs align the large first tap.



Turbo Equalization

- These are packet adaptive equalizers where L16:26's channel identification (of H) or partial-response equalization (L17:13) is used.
- A MLSE (Viterbi Detector) for the channel ISI is used in stead of the feedback section.



- The channel's memory is treated like a code with the SOVA generation of soft information



The intrinsic channel information

- Initially, Viterbi/SOVA produces ratios:

- Sum of such terms if $M^v > 2$.
- Evaluate each stage 0/1 among survivors.

$$\frac{e^{-\frac{1}{2\sigma^2} \cdot \|\mathbf{y} - \mathbf{H} \cdot \mathbf{x}_{k,0}\|^2}}{e^{-\frac{1}{2\sigma^2} \cdot \|\mathbf{y} - \mathbf{H} \cdot \mathbf{x}_{k,1}\|^2}}$$

- Later runs

- Include the code's soft extrinsic information in the Viterbi partial-response updates.

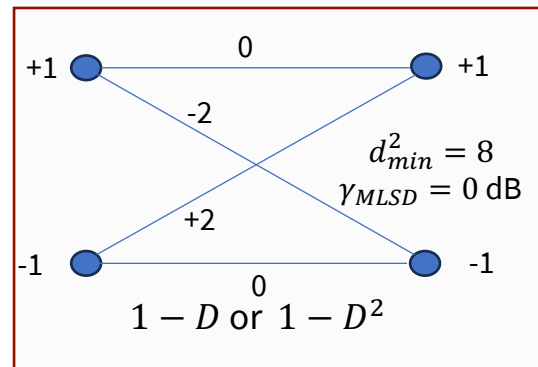
- The MLSD on channel trellis is optimum – lower initial P_e

- But loses advantage as number of levels increase in PAM/QAM
- Precoder can reduce this loss, but not eliminate it.

- The code and channel may interleave order w.r.t. each other.

- The SNR_{mf} attained by Viterbi does not

- Tends to prevent transmit-filter optimization.



**Much better to use
Decision Feedback
&
Good Code**

**Those can achieve
reliable
transmission at any
rate up to capacity**





End Lecture 17