



Lecture 12: Filter Types and Structures

Outlines

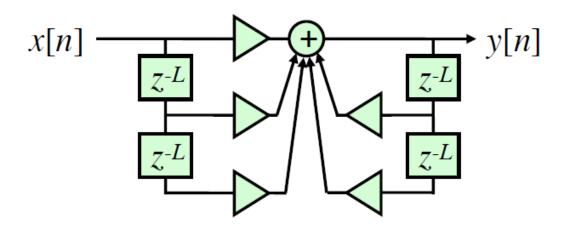
- 1. More filter types
- 2. Minimum and maximum phase
- 3. Filter implementation structures

1. More Filter Types

- We have seen the basics of filters and a range of simple examples
- Now look at a couple of other classes:
 - Comb filters multiple pass/stop bands
 - Allpass filters only modify signal phase

Comb Filters

Replace all system delays z⁻¹ with longer delays z^{-L}



→ System that behaves 'the same' at a longer timescale

Comb Filters

 'Parent' filter impulse response h[n] becomes comb filter output as:

$$g[n] = \{h[0] \ 0 \ 0 \ 0 \ h[1] \ 0 \ 0 \ 0 \ h[2]..\}$$

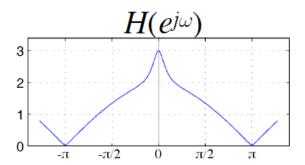
• Thus,
$$G(z) = \sum_{n} g[n]z^{-n}$$

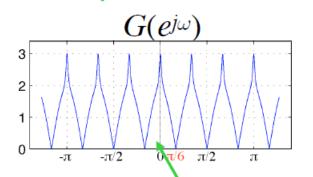
= $\sum_{n} h[n]z^{-nL} = H(z^{L})$

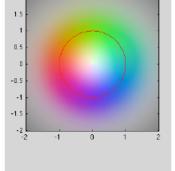
Comb Filters

Hence frequency response:

$$G(e^{j\omega}) = H(e^{j\omega L})$$
 parent frequency response compressed & repeated L times







- Low-pass response →
- L copies of $H(e^{j\omega})$
- pass $\omega = 0, 2 \pi/L, 4 \pi/L...$

• cut $\omega = \pi/L$, 3 π/L , 5 π/L ...

useful to enhance a harmonic series

- Allpass filter has $|A(e^{j\omega})|^2 = K$ for all ω i.e. spectral energy is not changed
- Phase response is not zero (else trivial)
 - phase correction
 special effects

e.g.

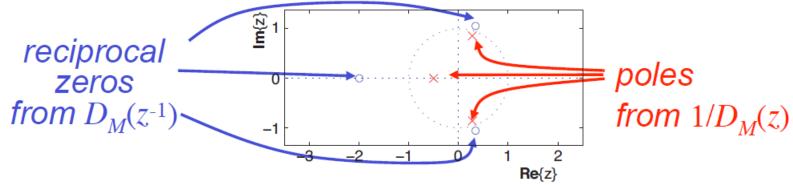
Allpass has special form of system fn:

$$\begin{split} A_{M}\left(z\right) &= \pm \frac{d_{M} + d_{M-1}z^{-1} + \ldots + d_{1}z^{-(M-1)} + z^{-M}}{1 + d_{1}z^{-1} + \ldots + d_{M-1}z^{-(M-1)} + d_{M}z^{-M}} \\ &= \pm z^{-M} \frac{D_{M}\left(z^{-1}\right)}{D_{M}\left(z\right)} = & \text{mirror-image polynomials} \end{split}$$

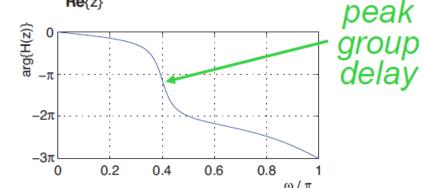
- $A_M(z)$ has poles λ where $D_M(\lambda) = 0$
 - $\rightarrow A_M(z)$ has zeros $\zeta = 1/\lambda = \lambda^{-1}$

$$A_M(z) = \pm z^{-M} \frac{D_M(z^{-1})}{D_M(z)}$$

• Any (stable) D_M can be used:



- Phase is always decreasing:
- \rightarrow -M π at $\omega = \pi$



Why do mirror-img poly's give const gain?

Conj-sym system fn can be factored as:

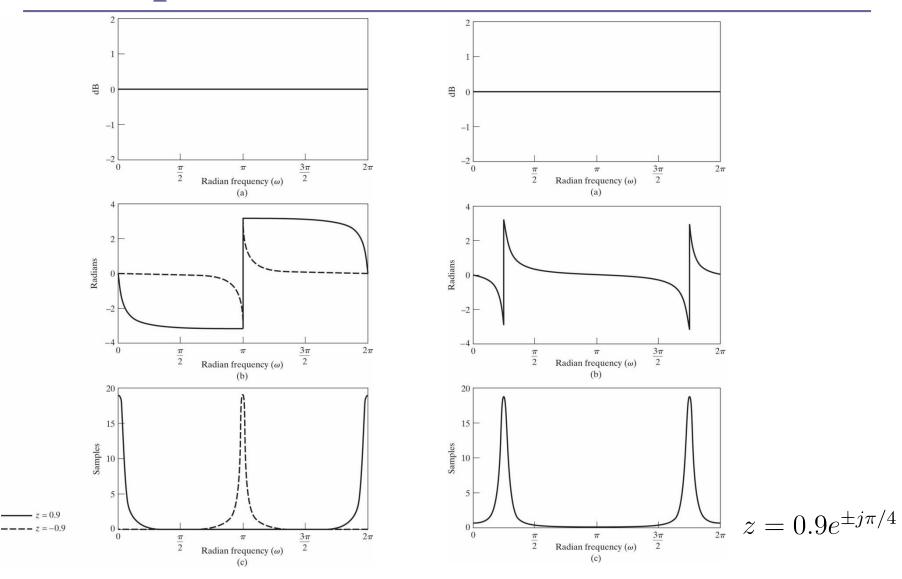
$$A_{M}(z) = \frac{K \prod_{i} (z - \lambda_{i}^{*-1})}{\prod_{i} (z - \lambda_{i})}$$

$$= \frac{K \prod_{i} \lambda_{i}^{*-1} z \left(\lambda_{i}^{*} - z^{-1}\right)}{\prod_{i} (z - \lambda_{i})}$$

$$= \frac{K \prod_{i} \lambda_{i}^{*-1} z \left(\lambda_{i}^{*} - z^{-1}\right)}{\prod_{i} (z - \lambda_{i})}$$
+ complex conjugate p/z

• $z = e^{j\omega} \rightarrow z^{-1} = e^{-j\omega}$ also on u.circle...

Example:

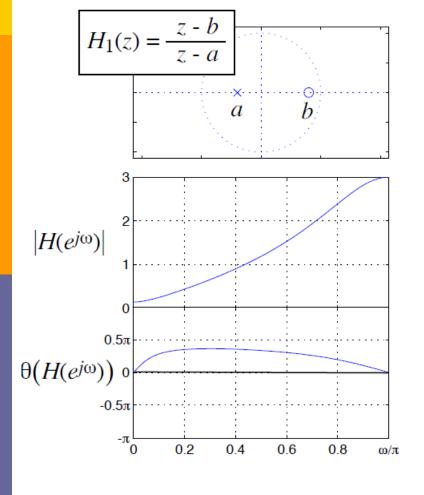


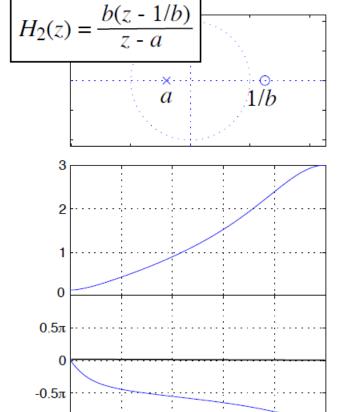
2. Minimum/Maximum Phase

- In AP filters, reciprocal roots have...
 - same effect on magnitude (modulo const.)
 - different effect on phase
- In normal filters, can try substituting reciprocal roots
 - reciprocal of stable pole will be unstable X
 - reciprocals of zeros?
- → Variants of filters with same magnitude response, different phase

Minimum/Maximum Phase

Hence:





0.4

8.0

ω/π

0.6

 $-\pi_0$

0.2

reciprocal zero..

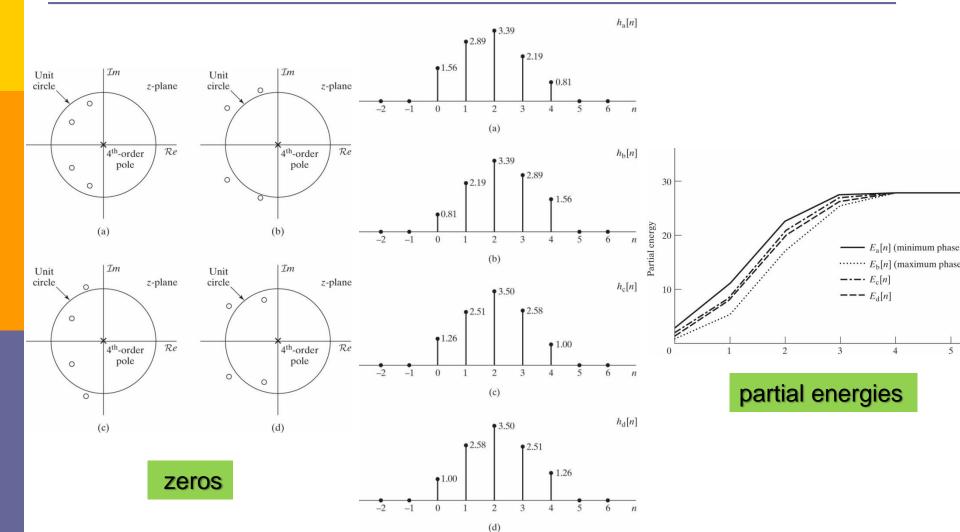
.. same mag..

.. added phase lag

Minimum/Maximum Phase

- For a given magnitude response
 - All zeros inside u.circle → minimum phase
 - All zeros outside u.c. → maximum phase (greatest phase dispersion for that order)
 - Otherwise, mixed phase
- i.e. for a given magnitude response several filters & phase fns are possible; minimum phase is canonical, 'best'

Example:

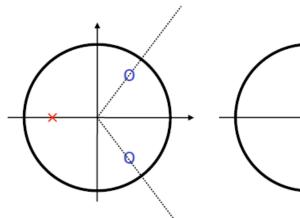


sequences

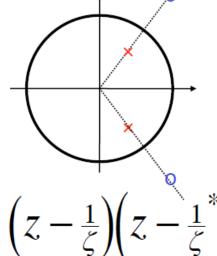
Minimum/Maximum Phase

Note:

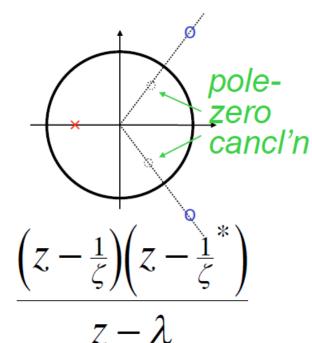
Min. phase + Allpass = Max. phase



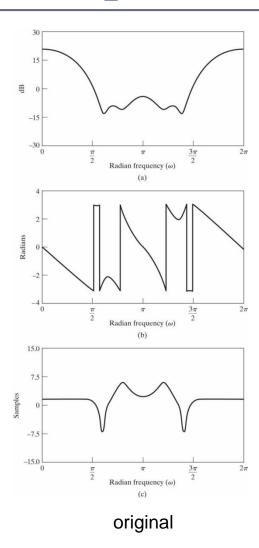
$$\frac{(z-\zeta)(z-\zeta^*)}{z-\lambda}$$



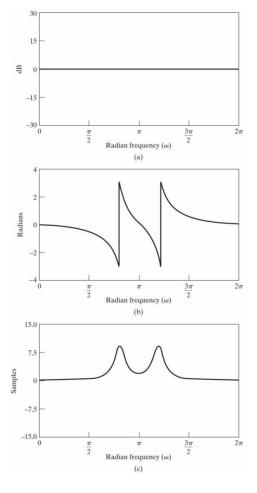
$$\frac{(z-\bar{\zeta})(z-\bar{\zeta})}{(z-\zeta)(z-\zeta^*)}$$



Example:



g o -15 -30 L $\frac{3\pi}{2}$ Radian frequency (ω) (a) Radians Radian frequency (ω) 7.5 Samples -15.0 L 2π Radian frequency (ω)



minimum-phase system

all-pass system

Inverse Systems

- $h_i[n]$ is called the inverse of $h_f[n]$ iff $h_i[n] \circledast h_f[n] = \delta[n]$
- **Z**-transform: $H_f(e^{j\omega}) \cdot H_i(e^{j\omega}) = 1$

$$\begin{array}{c|c}
x[n] & y[n] \\
\hline
 & H_i(z) & w[n]
\end{array}$$

$$W(z) = H_i(z)Y(z) = H_i(z)H_f(z)X(z) = X(z)$$
$$\Rightarrow w[n] = x[n]$$

• i.e. $H_i(z)$ recovers x[n] from o/p of $H_f(z)$

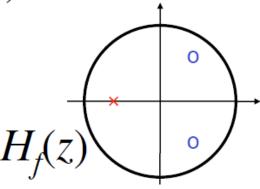
Inverse Systems

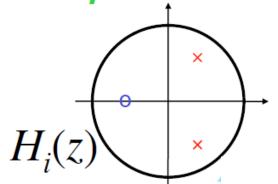
• What is $H_i(z)$? $H_i(z)H_f(z)=1$ $\Rightarrow H_i(z)=1/H_f(z)$

• $H_i(z)$ is reciprocal polynomial of $H_f(z)$

$$H_f(z) = \frac{P(z)}{D(z)} \Rightarrow H_i(z) = \frac{D(z)}{P(z)} \xrightarrow{\text{poles of fwd}} \frac{\text{poles of fwd}}{\text{zeros of fwd}}$$

Just swap poles and zeros:





Inverse Systems

When does $H_i(z)$ exist?

- Causal+stable \rightarrow all $H_i(z)$ poles inside u.c.
 - \rightarrow all zeros of $H_f(z)$ must be inside u.c.
 - $\rightarrow H_f(z)$ must be minimum phase
- $H_f(z)$ zeros outside u.c. → unstable $H_i(z)$
- $H_f(z)$ zeros **on** u.c. \rightarrow unstable $H_i(z)$

$$H_i(e^{j\omega}) = 1/H_f(e^{j\omega}) = 1/0|_{\omega=\zeta}$$

 \rightarrow only invert if min.phase, $\Rightarrow H_f(e^{j\omega}) \neq 0$

System Identification

$$\begin{array}{c}
x[n] \\
 \hline
 H(z)
\end{array}$$

- Inverse filtering = given y and H, find x
- System ID = given y (and $\sim x$), find H
- Just run convolution backwards?

$$y[n] = \sum_{k=0}^{\infty} h[k]x[n-k]$$

$$\Rightarrow y[0] = h[0]x[0] \rightarrow h[0]$$

$$\Rightarrow y[1] = h[0]x[1] + h[1]x[0] \rightarrow h[1]...$$

$$deconvolution$$

$$but: errors$$

$$accumulate$$

$$y[1] = h[0]x[1] + h[1]x[0] \rightarrow h[1]...$$

System Identification

$$x[n]$$
 $y[n] + noise$

Better approach uses correlations;
Cross-correlate input and output:

$$r_{xy}[\ell] = y[\ell] \circledast x[-\ell] = h_{?}[\ell] \circledast x[\ell] \circledast x[-\ell]$$
$$= h_{?}[\ell] \circledast r_{xx}[\ell]$$

- If r_{xx} is 'simple', can recover $h_{?}[n]...$
- e.g. (pseudo-) white noise:

$$r_{xx}[\ell] \approx \delta[\ell] \implies h_{?}[n] \approx r_{xy}[\ell]$$

System Identification

Can also work in frequency domain:

$$S_{xy}(z) = H_2(z) \cdot S_{xx}(z) \longleftarrow$$
 make a const.

• x[n] is not observable $\to S_{xy}$ unavailable, but $S_{xx}(e^{j\omega})$ may still be known, so:

$$S_{yy}(e^{j\omega}) = Y(e^{j\omega})Y^*(e^{j\omega})$$

$$= H(e^{j\omega})X(e^{j\omega})H^*(e^{j\omega})X^*(e^{j\omega})$$

$$= |H(e^{j\omega})|^2 \cdot S_{--}(e^{j\omega})$$

• Use e.g. min.phase to rebuild $H(e^{j\omega})$...

3. Filter Structures

- Many different implementations, representations of same filter
- Different costs, speeds, layouts, noise performance, ...

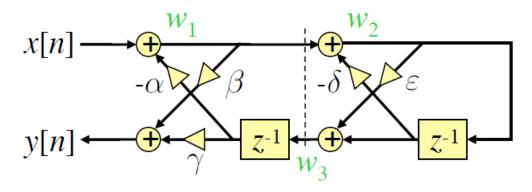
Block Diagrams

- Useful way to illustrate implementations
- Z-transform helps analysis:

- Output of summers as dummy variables
- Everything else is just multiplicative

Block Diagrams

More complex example:



$$Y = \gamma z^{-1} W_3 + \beta W_1$$

$$\Rightarrow \frac{Y}{X} = \frac{\beta + z^{-1} (\beta \delta + \gamma \varepsilon) + z^{-2} (\gamma)}{1 + z^{-1} (\delta + \alpha \varepsilon) + z^{-2} (\alpha)}$$

$$\xrightarrow{\text{stackable}}_{\text{2nd order section}}$$

$$W_{1} = X - \alpha z^{-1} W_{3}$$

$$W_{2} = W_{1} - \delta z^{-1} W_{2}$$

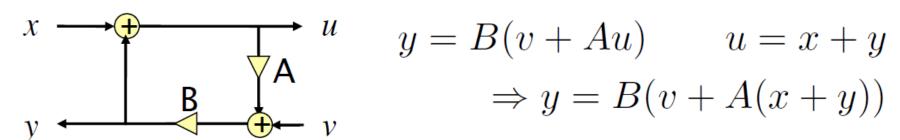
$$W_{3} = z^{-1} W_{2} + \varepsilon W_{2}$$

$$W_{2} = \frac{W_{1}}{1 + \delta z^{-1}}$$

$$W_{3} = \frac{(z^{-1} + \varepsilon)W_{1}}{1 + \delta z^{-1}}$$

Delay-Free Loops

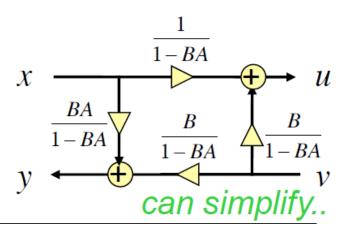
Can't have them!



- At time n = 0, setup inputs x and y; need u for y, also y for $u \rightarrow can't$ calculate
- Algebra:

$$y(1 - BA) = Bv + BAx$$

$$\Rightarrow y = \frac{Bv + BAx}{1 - BA}$$

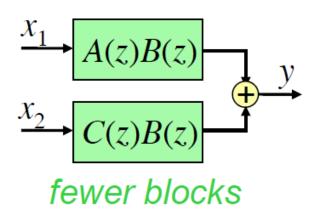


Equivalent Structures

- Modifications to block diagrams that do not change the filter
- e.g. Commutation H = AB = BA

$$\longrightarrow A \longrightarrow B \longrightarrow B \longrightarrow A \longrightarrow$$

• Factoring $AB+CB=(A+C)\cdot B$



$$X_1 \longrightarrow A(z)$$

$$X_2 \longrightarrow C(z)$$

$$B(z) \longrightarrow B(z)$$

less computation

Equivalent Structures

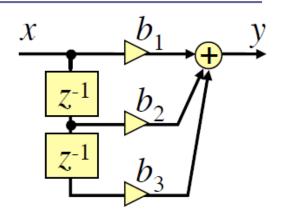
Transpose

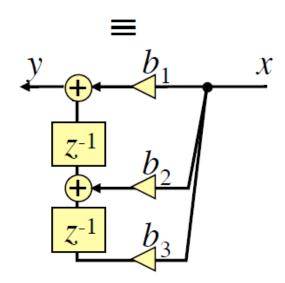
- reverse paths
- adders

 nodes
- input → output

$$Y = b_1 X + b_2 z^{-1} X + b_3 z^{-2} X$$

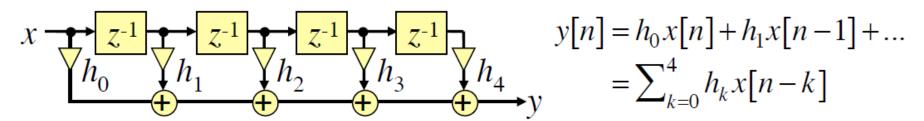
= $b_1 X + z^{-1} (b_2 X + z^{-1} b_3 X)$



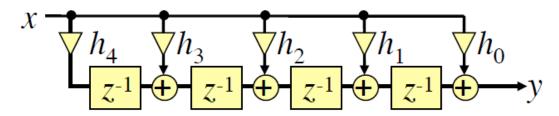


FIR Filter Structures

Direct form "Tapped Delay Line"



Transpose



Re-use delay line if several inputs x_i for single output y?

FIR Filter Structures

Cascade

factored into e.g. 2nd order sections

$$H(z) = h_0 + h_1 z^{-1} + h_2 z^{-2} + h_3 z^{-3}$$

$$= h_0 \left(1 - \zeta_0 z^{-1} \right) \left(1 - \zeta_1 z^{-1} \right) \left(1 - \zeta_1^* z^{-1} \right)$$

$$= h_0 \left(1 - \zeta_0 z^{-1} \right) \left(1 - 2 \operatorname{Re} \{ \zeta_1 \} z^{-1} + |\zeta_1|^2 z^{-2} \right)$$

$$\xrightarrow{x \quad h_0 \quad y}$$

$$\xrightarrow{z^{-1} \quad \zeta_0 \quad z^{-1} \quad |\zeta_1|^2}$$

FIR Filter Structures

Linear Phase:

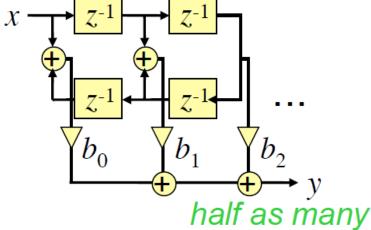
$$\frac{1}{\sqrt{1+\frac{1}{2}}} \frac{1}{\sqrt{1+\frac{1}{2}}} \frac{1}{\sqrt{1+\frac{$$

Symmetric filters with h[n] = (-)h[N - n]

$$y[n] = b_0(x[n] + x[n-4])$$

$$+b_1(x[n-1] + x[n-3])$$

$$+b_2x[n-2]$$



Also Transpose form: multiplies
 gains first, feeding folded delay/sum line

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IIR Filter Structures

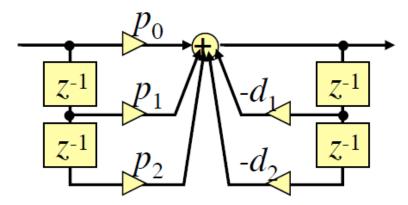
IIR: numerator + denominator

$$H(z) = \frac{p_0 + p_1 z^{-1} + p_2 z^{-2} + \dots}{1 + d_1 z^{-1} + d_2 z^{-2} + \dots}$$

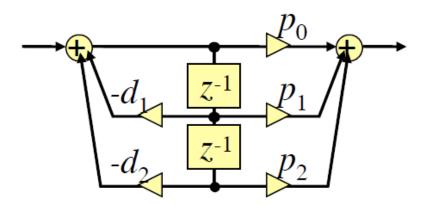
$$= P(z) \cdot \frac{1}{D(z)}$$
FIR
$$\frac{p_0}{z^{-1}} = \frac{p_1}{p_2}$$
all-pole IIR

IIR Filter Structures

Hence, Direct form I



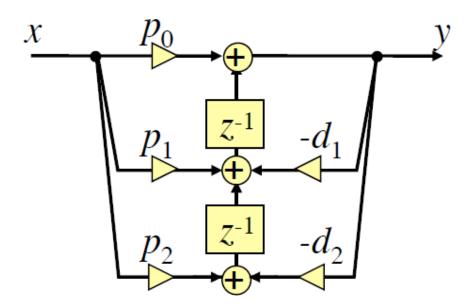
Commutation → Direct form II (DF2)



- same signaldelay lines merge
- "canonical"= min. memory usage

IIR Filter Structures

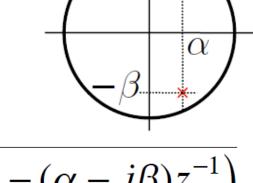
Use Transpose on FIR/IIR/DF2



"Direct Form II Transpose"

Factored IIR Structures

Real-output filters have conjugate-symm roots:



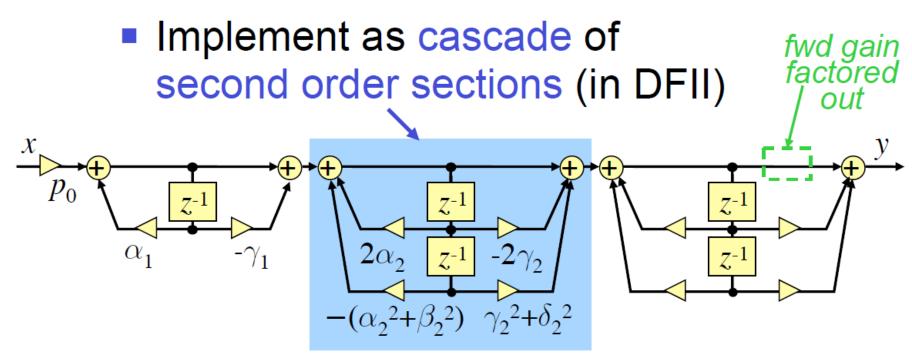
$$H(z) = \frac{1}{(1 - (\alpha + j\beta)z^{-1})(1 - (\alpha - j\beta)z^{-1})}$$

Can always group into 2nd order terms with real coefficients:

$$H(z) = \frac{p_0 \left(1 - \gamma_1 z^{-1}\right) \left(1 - 2\gamma_2 z^{-1} + (\gamma_2^2 + \delta_2^2) z^{-2}\right) \dots}{\left(1 - \alpha_1 z^{-1}\right) \left(1 - 2\alpha_2 z^{-1} + (\alpha_2^2 + \beta_2^2) z^{-2}\right) \dots}$$

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Cascade IIR Structures



- Second order sections (SOS):
 - modular any order from optimized block
 - well-behaved, real coefficients (sensitive?)

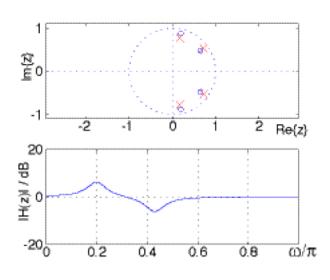
Second-Order Sections

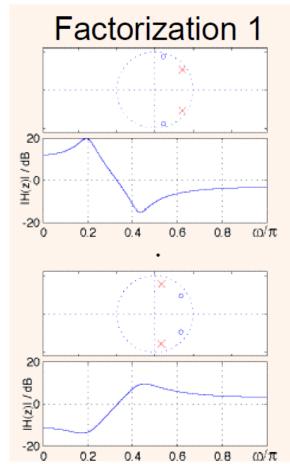
- 'Free' choices:
 - grouping of pole pairs with zero pairs
 - order of sections
- Optimize numerical properties:
 - avoid very large values (overflow)
 - avoid very small values (quantization)
- e.g. Matlab's zp2sos
 - attempt to put 'close' roots in same section
 - intersperse gain & attenuation?

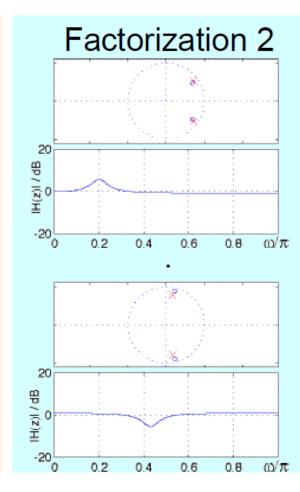
Second-Order Sections

Factorization affects intermediate values

Original System (2 pair poles, zeros)







Parallel IIR Structures

• Can express H(z) as sum of terms (IZT)

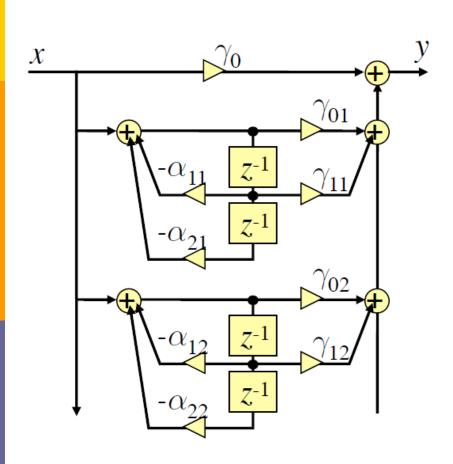
$$H(z) = \operatorname{consts} + \sum_{\ell=1}^{N} \frac{\rho_{\ell}}{1 - \lambda_{\ell} z^{-1}} \qquad \rho_{\ell} = (1 - \lambda_{\ell} z^{-1}) F(z)|_{z = \lambda_{\ell}}$$

Or, second-order terms:

$$H(z) = \gamma_0 + \sum_{k} \frac{\gamma_{0k} + \gamma_{1k} z^{-1}}{1 + \alpha_{1k} z^{-1} + \alpha_{2k} z^{-2}}$$

Suggests parallel realization...

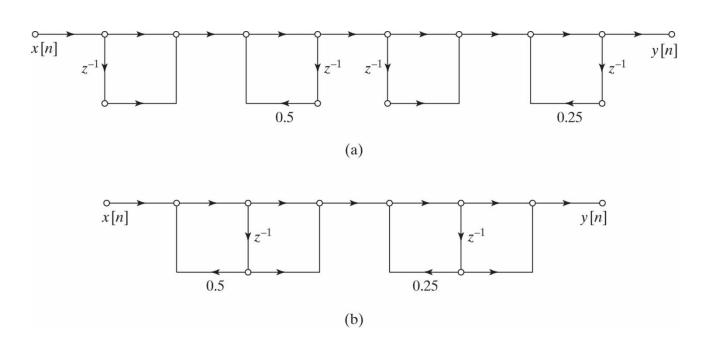
Parallel IIR Structures



- Sum terms become parallel paths
- Poles of each SOS are from full TF
- System zeros arise from output sum
- Why do this?
 - stability/sensitivity
 - reuse common terms

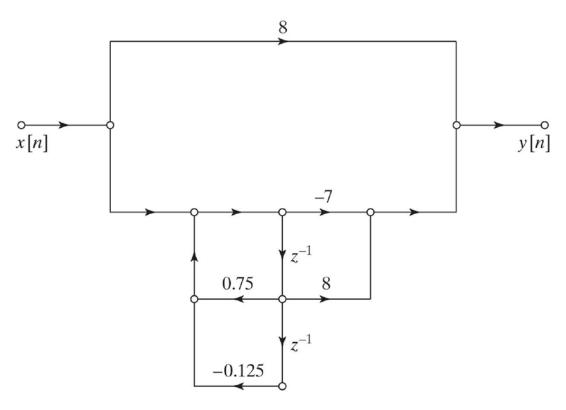
Example – Cascade Structure

$$H(z) = \frac{1 + 2z^{i-1} + z^{i-2}}{1i \cdot 0:75z^{i-1} + 0:125z^{i-2}} = \frac{(1 + z^{i-1})(1 + z^{i-1})}{(1i \cdot 0:5z^{i-1})(1i \cdot 0:25z^{i-1})}$$



Example – Parallel Structure

$$H(z) = \frac{1 + 2z^{i-1} + z^{i-2}}{1i \cdot 0.75z^{i-1} + 0.125z^{i-2}} = 8 + \frac{i \cdot 7 + 8z^{i-1}}{1i \cdot 0.75z^{i-1} + 0.125z^{i-2}}$$



Example – Parallel Structure (Cont.)

$$H(z) = 8 + \frac{18}{1 \text{ i } 0.5z^{\text{i } 1}} + \frac{25}{1 \text{ i } 0.25z^{\text{i } 1}}$$

