EE 508 Lecture 16

Filter Transformations

Lowpass to Highpass Lowpass to Band-reject

Filter Synthesis

Standard LP to BP Transformation $s \rightarrow \frac{s^2+1}{s \cdot BW_{N}}$

- Standard LP to BP transform is a variable mapping transform
- Maps j ω axis to j ω axis
- Maps LP poles to BP poles
- Preserves basic shape but warps frequency axis
- Doubles order
- Pole Q of resultant band-pass functions can be very large for narrow pass-band
- Sequencing of frequency scaling and transformation does not affect final function

Review from Last Time Standard LP to BP Transformation



Review from Last Time Standard LP to BP Transformation

Frequency and s-domain Mappings - Denormalized

(subscript variable in LP approximation for notational convenience)



All three approaches give same approximation

Which is most practical to use?

Often none of them !

Example 1: Obtain an approximation that meets the following specifications



BW= $\omega_{\rm B}$ - $\omega_{\rm A}$

$$\omega_{\rm M} = \sqrt{\omega_{\rm B} \bullet \omega_{\rm A}}$$

Assume that $\omega_{\text{AL}},\,\omega_{\text{BH}}$ and ω_{M} satsify

$$\frac{\omega_{\rm M}^2 \textbf{-} \omega_{\rm AL}^2}{\omega_{\rm AL}} \textbf{-} \frac{\omega_{\rm BH}^2 \textbf{-} \omega_{\rm M}^2}{\omega_{\rm BH}} \textbf{-} \frac{\omega_{\rm BH}^2 \textbf{-} \omega_{\rm M}^2}{\omega_{\rm BH}}$$

Example 1: Obtain an approximation that meets the following specifications



(actually ω_A and ω_{AL} that map to -1 and - ω_s respectively but show 1 and ω_s for convenience)

Example 2: Obtain an approximation that meets the following specifications





 $\omega_{_{SN}} = \min\{\omega_{_{S1}}, \omega_{_{S2}}\}$

Example 2: Obtain an approximation that meets the following specifications



Filter Transformations

Lowpass to Bandpass (LP to BP) Lowpass to Highpass (LP to HP) Lowpass to Band-reject (LP to BR)

- Approach will be to take advantage of the results obtained for the standard LP approximations
- Will focus on flat passband and zero-gain stop-band transformations

Flat Passband/Stopband Filters



LP to BS Transformation

Strategy: As was done for the LP to BP approximations, will use a variable mapping strategy that maps the imaginary axis in the s-plane to the imaginary axis in the s-plane so the basic shape is preserved.

$$\begin{array}{c} X_{IN} \longrightarrow \\ T_{LPN}(s) \longrightarrow \\ T_{BS}(s) = T_{LPN}(f(s)) \end{array} \begin{array}{c} X_{OUT} & X_{OUT} & X_{OUT} \\ T_{BS}(s) = T_{LPN}(f(s)) \end{array}$$

$$f(s) = \frac{\sum_{i=0}^{m_{T}} a_{Ti} s^{i}}{\sum_{i=0}^{n_{T}} b_{Ti} s^{i}}$$

m

LP to BS Transformation





Variable Mapping Strategy to Preserve Shape of LP function:

 $F_N(s)$ should

```
map s=0 to s=± j\infty
map s=0 to s= j0
map s=j1 to s=j\omega_A
map s=j1 to s=-j\omega_B
map s=-j1 to s=j\omega_B
map s=-j1 to s=-j\omega_A
```



map $\omega=0$ to $\omega = \pm \infty$ map $\omega=0$ to $\omega = 0$ map $\omega=1$ to $\omega = \omega_A$ map $\omega=1$ to $\omega = -\omega_B$ map $\omega = -1$ to $\omega = \omega_B$ map $\omega = -1$ to $\omega = -\omega_A$





Mapping Strategy: consider variable mapping transform

```
F_{N}(s) \text{ should}
map s=0 to s=\pm j\infty

map s=0 to s=j0

map s=j1 to s=j\omega_{A}

map s=j1 to s=-j\omega_{B}

map s=-j1 to s=-j\omega_{A}
```

map $\omega=0$ to $\omega = \pm \infty$ map $\omega=0$ to $\omega = 0$ map $\omega=1$ to $\omega = \omega_A$ map $\omega=1$ to $\omega = -\omega_B$ map $\omega = -1$ to $\omega = \omega_B$ map $\omega = -1$ to $\omega = -\omega_A$

Consider variable mapping

$$\begin{aligned} \mathsf{T}_{\mathsf{LPN}}\left(F_{N}(s)\right) &= \mathsf{T}_{\mathsf{BSN}}\left(s\right)\Big|_{s=\frac{\mathsf{s} \cdot \mathsf{BW}_{\mathsf{N}}}{\mathsf{s}^{2}+1}} \\ \mathsf{s} \to \frac{\mathsf{s} \cdot \mathsf{BW}_{\mathsf{N}}}{\mathsf{s}^{2}+1} \end{aligned}$$

Comparison of Transforms



Frequency and s-domain Mappings

(subscript variable in LP approximation for notational convenience)



Un-normalized Frequency and s-domain Mappings

(subscript variable in LP approximation for notational convenience)



Pole Mappings



Can show that the upper hp pole maps to one upper hp pole and one lower hp pole as shown. Corresponding mapping of the lower hp pole is also shown

- Poles lie on a constant-Q line
- Zeros at $\pm j1$ (normalized) or at $\pm j\omega_M$ (un-normalized) of multiplicity n

LP to BS Transformation



Note for γ small, Q_{BS} can get very large



Note doubling of poles, addition of zeros, and likely Q enhancement

$$s_x \rightarrow \frac{s \bullet BW}{s^2 + \omega_M^2}$$

- Standard LP to BS transformation is a variable mapping transform
- Maps j ω axis to j ω axis in the s-plane
- Preserves basic shape of an approximation but warps frequency axis
- Order of BS approximation is double that of the LP Approximation
- Pole Q and ω_0 expressions are identical to those of the LP to BP transformation
- Pole Q of BS approximation can get very large for narrow BW
- Other variable transforms exist but the standard is by far the most popular

Filter Transformations

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• Will focus on flat passband and zero-gain stop-band transformations

Flat Passband/Stopband Filters



LP to HP Transformation

Strategy: As was done for the LP to BP approximations, will use a variable mapping strategy that maps the imaginary axis in the s-plane to the imaginary axis in the s-plane so the basic shape is preserved.

$$X_{IN} \longrightarrow T_{LPN}(s) \xrightarrow{X_{OUT}} S \longrightarrow f(S) \qquad X_{IN} \longrightarrow T_{HP}(s) \xrightarrow{X_{OUT}} T_{HP}(s) \xrightarrow{X_{OUT}$$

LP to HP Transformation





Mapping Strategy:



Variable Mapping Strategy to Preserve Shape of LP function:

 $F_N(s)$ should

map s=0 to s= $\pm j \infty$ map s=j1 to s=-j1map s=-j1 to s=j1



map $\omega=0$ to $\omega=\infty$ map $\omega=1$ to $\omega=-1$ map $\omega=-1$ to $\omega=1$



Mapping Strategy: consider variable mapping transform

F_N(s) should

map s=0 to s= $\pm j\infty$ map s=j1 to s=-j1 map s= -j1 to s=j1



map $\omega=0$ to $\omega=\infty$ map $\omega=1$ to $\omega=-1$ map $\omega=-1$ to $\omega=1$

Consider variable mapping

Comparison of Transforms



LP to HP Transformation





Frequency and s-domain Mappings

(subscript variable in LP approximation for notational convenience)



Pole Mappings

Claim: With a variable mapping transform, the variable mapping naturally defines the mapping of the poles of the transformed function



Pole Mappings



Pole Mappings





Highpass poles are scaled in magnitude but make identical angles with imaginary axis

HP pole Q is same as LP pole Q

Order is preserved

(Un-normalized variable mapping transform)



Filter Design Process



There are many different filter architectures that can realize a given transfer function

Considerable effort has been focused over the years on "inventing" these architectures and on determining which is best suited for a given application

Most even-ordered designs today use one of the following three basic architectures



Multiple-loop Feedback (less popular)



What's unique in all of these approaches?

Most odd-ordered designs today use one of the following three basic architectures



Multiple-loop Feedback (less popular)



What's unique in all of these approaches?

What's unique in all of these approaches?



- Most effort on synthesis can focus on synthesizing these four blocks (the summing function is often incorporated into the Biquad or Integrator) (the first-order block is much less challenging to design than the biquad)
- Some issues associated with their interconnections
- And, in many integrated structures, the biquads are made with integrators (thus, much filter design work simply focuses on the design of integrators)

Biquads

How many biquad filter functions are there?



$$T(s) = \frac{a_2 s^2 + a_1 s + a_0}{s^2 + b_1 s + b_0} \qquad a_0 \neq 0, \ a_1 \neq 0, \ a_2 \neq 0$$

$$T(s) = \frac{a_0}{s^2 + b_1 s + b_0} \qquad a_0 \neq 0 \qquad T(s) = \frac{a_2 s^2 + a_0}{s^2 + b_1 s + b_0} \qquad a_0 \neq 0, \ a_2 \neq 0$$

$$T(s) = \frac{a_1 s}{s^2 + b_1 s + b_0} \qquad a_1 \neq 0 \qquad T(s) = \frac{a_1 s + a_0}{s^2 + b_1 s + b_0} \qquad a_0 \neq 0, \ a_1 \neq 0$$

$$T(s) = \frac{a_2 s^2}{s^2 + b_1 s + b_0} \qquad a_2 \neq 0 \qquad T(s) = \frac{a_2 s^2 + a_1 s}{s^2 + b_1 s + b_0} \qquad a_2 \neq 0, \ a_1 \neq 0$$

Review: Second-order bandpass transfer function



X_{IN___} X_{OUT} T(s)



There are many different filter architectures that can realize a given transfer function

Will first consider second-order Bandpass filter structures



There are many different filter architectures that can realize a given transfer function

Will first consider second-order Bandpass filter structures



Second-order Bandpass Filter

3 degrees of freedom

2 degrees of freedom for determining dimensionless transfer function (impedance values scale)

$$\omega_0 = ?$$
 Q = ?



Can realize an arbitrary stable 2^{nd} order bandpass function within a gain factor Simple design process (sequential but not independent control of ω_0 and Q)

If trimming is necessary, prefer to trim with a single resistor

Can't trim this filter with single resistor



Second-order Bandpass Filter

6 degrees of freedom (effectively 5 because dimensionless) Denote as a +KRC filter

$$\omega_0 = ?$$
 $Q = ? BW = ?$



 $\omega_0 = ?$ Q = ?

BW = ?



- Can satisfy arbitrary 2nd=order BP constraints within a gain factor with this circuit
 - Very simple circuit structure
 - Independent control of ω_0 and Q but requires tuning more than one component
- Can actually move poles in RHP by making K >4









Can't trim this filter with resistor



Second-order Bandpass Filter

5 degrees of freedom (4 effective since dimensionless)

Denote as a -KRC filter

 $\omega_0 = ?$ Q = ?

BW = ?



3 degrees of freedom

 $\omega_0 = ?$ Q = ? BW = ?



3 degrees of freedom (2 effective since dimensionless)

- Can satisfy arbitrary 2nd=order BP constraints within a gain factor with this circuit
- Very simple circuit structure
- Simple design process (sequential but not independent control of ω_0 and Q, requires tuning of more than 1 component if Rs used)

Observation:



These are often termed Sallen and Key filters

Sallen and Key introduced a host of filter structures

Sallen and Key structures comprised of summers, RC network, and finite gain amplifiers

These filters were really ahead of their time and appeared long before practical implementations were available

IRE TRANSACTIONS-CIRCUIT THEORY

March 1955

A Practical Method of Designing RC Active Filters*

R. P. SALLEN[†] and E. L. KEY[†]



Stay Safe and Stay Healthy !

End of Lecture 16