ECE 5180/6180 Microwave Filter Design

Lectures:

Lumped element filters (also applies to low frequency filters) Stub Filters Stepped Impedance Filters Coupled Line Filters

Lumped Element Filters

Text Section 8.3

Portfolio Question: How do you design a lumped element filter using the Insertion Loss Method

Power Loss Ratio

 $P_{LR} = \frac{Power \ Available \ from \ Source}{P_{out} \ Delivered \ to \ Load} = \frac{P_{inc}}{P_{load}}$ Insertion Loss

$$IL = 10 \ log \ P_{LR}$$

2-port network:

$$V_{1}^{+} = \begin{bmatrix} S_{11} & S_{12} \\ S_{21} & S_{22} \end{bmatrix} \begin{bmatrix} V_{1}^{+} \\ V_{2}^{-} \end{bmatrix} = \begin{bmatrix} S_{11} & S_{12} \\ S_{21} & S_{22} \end{bmatrix} \begin{bmatrix} V_{1}^{+} \\ V_{2}^{+} \end{bmatrix}$$

$$Psource = \frac{(V_{1}^{+})^{2}}{Z_{in}} \quad Pload = \frac{(V_{2}^{-})^{2}}{Z_{L}}$$

$$For \; Matched \; System \; Z_{in} = Z_{L} = Z_{o}$$

$$Then$$

$$\frac{Psource}{Pload} = \left(\frac{V_{1}^{+}}{V_{2}^{-}}\right)^{2} = \frac{1}{|S_{21}|^{2}} = \frac{1}{|S_{12}|^{2}}$$

$$(For \; \text{Re } ciprocal \; System \; S_{21} = S_{12})$$

$$P_{LR} = \frac{1}{1 - |\Gamma_{in}|^{2}} = \frac{1}{1 - |\Gamma_{in}(\omega)|^{2}}$$

Filter Design by insertion loss method controls $\Gamma(\omega)$ to control passband and stopband of filter.

Filter parameters:

Passband -- frequencies that are passed by filter Stopband -- frequencies that are rejected Insertion loss -- how much power is transferred to load in passband Attenuation -- how much power is rejected (not transferred to the load) in the stopband Cutoff rate or attenuation rate -- how quickly the filter transitions from pass-to-stop or stop-to-passbands

Phase response -- Linear phase response in the passband means that signal will not be distorted.

Classes of Filters:

Determined by Γ .

We have not proven this yet, but a useful mathematical proof (section 4.1) shows that $|\Gamma(\omega)|^2$ is an even function of ω . So $|\Gamma(\omega)|^2$ can be written as a polynomial in ω^2 .

$$\left|\Gamma(\omega)\right|^{2} = \frac{M(\omega^{2})}{M(\omega^{2}) + N(\omega^{2})}$$
$$P_{LR} = 1 + \frac{M(\omega^{2})}{N(\omega^{2})}$$

The class of filter is controlled by the type of polynomial used. Polynomials M and N can be

- Binomial (Butterworth) -- Maximally Flat
- Chebyshev -- Equal Ripple
- Elliptic -- Specified Minimum Stopband Attenuation (faster cutoff)
- Linear Phase

Binomial / Butterworth / Maximally Flat

Low Pass Filter Design:

$$P_{LR} = 1 + k^2 \left(\frac{\omega}{\omega_c}\right)^{2N}$$

$$\begin{split} N &= Filter \ Order \\ \omega &= frequency \ of \ interest \\ \omega_c &= cutoff \ frequency \end{split}$$

At ω_c , $P_{LR} = 1+k^2$ If the -3dB point is defined to be the cutoff point (common), k=1 For $\omega >> \omega_c$ then $P_{LR} \approx k^2 (\omega/\omega_c)^{2N}$ which means Insertion Loss increases at a rate of 20N dB / decade. (This allows us to increase the steepness of the cutoff by adding more sections.)

Chebyshev / Equal Ripple Filters

$$P_{LR} = 1 + k^2 T_N^2 \left(\frac{\omega}{\omega_c}\right)$$

Where T_N are Chebyshev polynomials

Ripples are equal-in-size = $1+k^2$ Cutoff Rate is 20N dB/decade, same as binomial. Insertion loss in the stopband is $(2^{2N})/4$ greater than binomial.

Elliptic and Linear Phase Filters

Other options, see textbook.

Filter Design Method

- 1. Design a LP filter for normalized Z,ω
- 2. Scale Z.

- 3. Convert from LP to HP or BP as desired.
- 4. Convert from lumped to distributed elements as desired.

1. Binomial Design of LP Filter for Normalized Z,ω

a. Determine how many elements are needed (N) Find ω/ω_c and look at the figure for attenuation in the stopband (Fig.8.26, p.450)

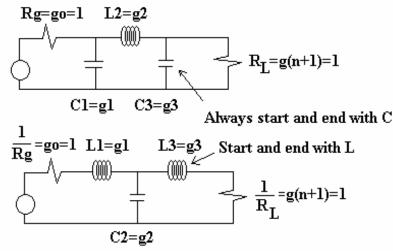
Example: How many elements are required to design a maximally-flat filter with a cutoff frequency of 2 GHz if the filter must provide 20 dB of attenuation at 4 GHz?

For this case, $|\omega/\omega_c|-1 = |4/2| - 1 = 1.0$ (bottom axis). Find N line on filter that is ABOVE the desired attenuation. N=4.

b. Find resistance or conductance values from Table 8.3

Look at N=3. g1 = 1.0; g2 = 2.0; g3 = 1.0; g4 = 1.0

c. Choose LP Filter Prototype Why choose one over the other? Available components. (Responses of both are identical.)



Notes:

- 1) Rg and RL must be REAL. What if they aren't? (Add a length of line, resonate, or absorb imaginary part.)
- 2) The designs in our book always have Rg=RL. What if they are not equal? There are other tables... see handout. This is effectively matching and filtering simultaneously.
- 3) Design so far has considered normalized impedances Rg=RL=1 and normalized frequency ($\omega_c = 1$) ... Use impedance and frequency scaling if they aren't 1.

1. Chebyshev (Equal Ripple) Design of LP Filter for Normalized Z, ω

Same steps as for binomial. There are only a few differences....

- (a) Determine number of elements (N). This will always be less than or equal to binomial. Use Figure 8.27, with choice of size of ripple.
- (b) Use table 8.4, with same choice of ripple as used in part a.
- (c) Same as binomial.

2. Impedance and Frequency Scaling (normalization)

To build the same filter for Zo = Rg = RL and a given cutoff frequency ω_c Use the same filter prototypes but scale the values:

- - $\begin{array}{c} Rg=1/(Zo\ go) \\ L1=(Zo)(g1) / \ \omega_c \\ C2=g2 / (Zo\ \omega_c) \\ L3=(Zo)(g3) / \ \omega_c \\ RL=1/(Zo\ g4) \end{array}$

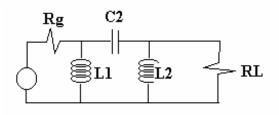
Binomial and Chebyshev are the same here, except for one difference: RL for binomial is always matched. RL for odd-order Chebyshev filters is NOT matched. Use a quarter-wave transformer to match.

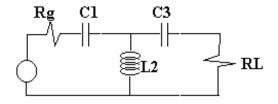
Example:

For 0.5 dB ripple and N=3, g3=1.9841. For top prototype, RL=1.9841 Zo, which is not matched. Quarterwave transformer would have $Zq=(\sqrt{1.9841})$ Zo

3. Convert from LP to HP

High Pass Configurations





$$\label{eq:constraint} \begin{split} & For \ both \ prototypes: \\ & C_k = 1/ \ (Zo \ \omega_c \ g_k) \\ & L_k = Zo \ (\omega_c \ g_k) \\ & For \ top \ prototype: \\ & Rg = Zo \ go \\ & R_L = Zo \ / \ g_{n+1} \\ & For \ bottom \ prototype: \\ & Rg = 1 \ / \ (Zo \ g_0) \\ & R_L = 1/ \ (Zo \ g_{n+1}) \end{split}$$

3. Convert from LP to Bandpass or Bandstop

Normalized bandwidth

 Δ

$$= \frac{\omega_2 - \omega_1}{\omega_o}$$

$$\omega_1 = \text{lower limit}$$

$$\omega_2 = \text{upper limit}$$

$$\omega_0 = \sqrt{\omega_1 \omega_2}$$

See Table 8.6 p. 461 for conversions.

To use for unnormalized filters:

 $L \rightarrow L Z_0$ $C \rightarrow C / Z_0$