6.5 Filter Design

Module:6 Microwave Passive circuits

Course: BECE305L – Antenna and Microwave Engineering

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Module:6 Microwave Passive circuits <u>7</u> hours

• T junction and resistive power divider, Wilkinson power divider, branch line coupler (equal & unequal), Rat Race Coupler, Filter design: Low pass filter (Butterworth and Chebyshev) - Richards transformation and stepped impedance methods.

Source of the contents: Pozar

• Filter is a two-port network

used to control the frequency response at a certain point in an RF or microwave system

by providing transmission at frequencies within the passband of the filter

and

attenuation in the stopband of the filter.

• Typical frequency responses (characteristics)

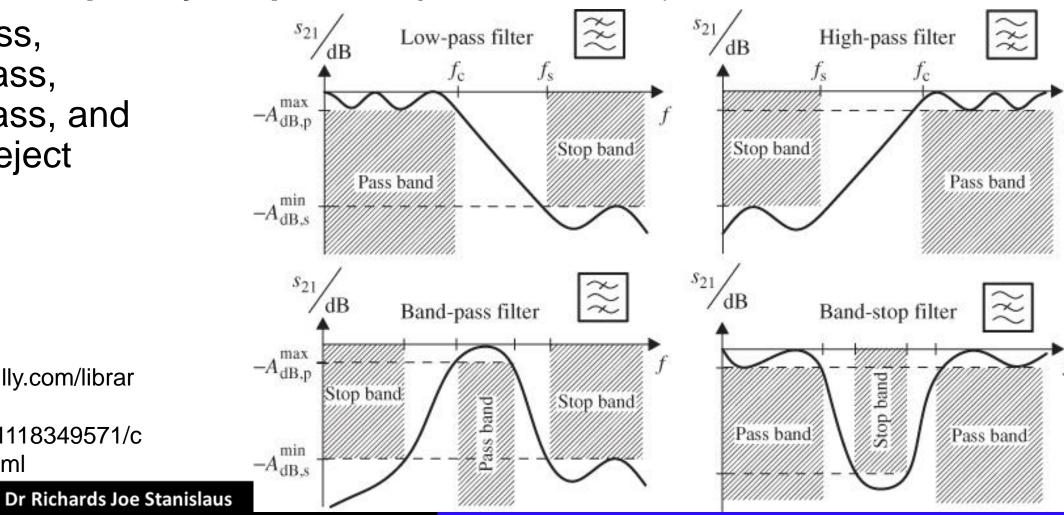
low-pass, high-pass, bandpass, and band-reject

Source: https://www.oreilly.com/librar y/view/rf-andmicrowave/9781118349571/c 06_level1_4.xhtml

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The design is **simplified** by beginning with **low-pass filter prototypes that are normalized in terms of impedance and frequency**.

Transformations are then applied to convert the prototype designs to the desired frequency range and impedance level.

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- Such filters do not exist in practice,
- so compromises must be made;

 The insertion loss method, however, allows a high degree of control over the passband and stopband amplitude and phase characteristics, with a systematic way to synthesize a desired response.

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- If, for example, a minimum insertion loss is most important, a binomial response could be used; a Chebyshev response would satisfy a requirement for the sharpest cutoff.
 - If it is possible to sacrifice the attenuation rate, a better phase response can be obtained by using a linear phase filter design.

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- the insertion loss method allows filter performance to be improved in a straightforward manner, at the expense of a higher order filter.
- For the filter prototypes to be discussed below, the order of the filter is equal to the number of reactive elements.

• Insertion loss method: a filter response is defined by its insertion loss, or power loss ratio, P_{LR} :

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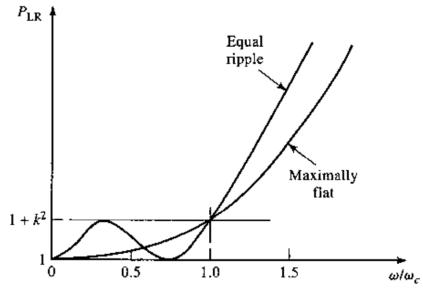
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- $P_{LR} = 1 + \frac{M(\omega^2)}{N(\omega^2)}$ Reflection ratio is also constrained.

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· Low pass filter, it is specified by

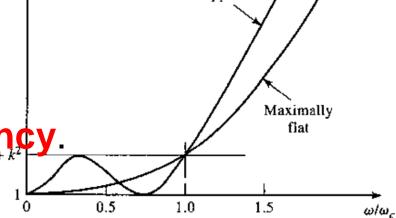
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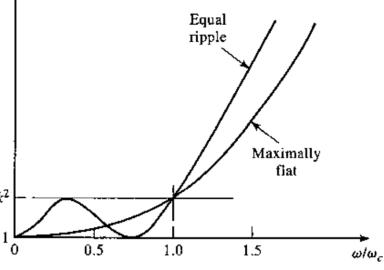
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- Pass band extends from $\omega=0$ to $\omega=\omega_c$
- At the edge, the power loss ratio is $1+k^2$ (If -3dB point, k=1, $P_{LR}=1+1=2=\frac{P_{inc}}{P_{load}}$)



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- Like binomial response, for multisection quarter-wave matching transformer, the first (2N-1) derivatives are zero at $\omega=0$.

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Chebyshev case's Insertion loss is $> \frac{2^{2N}}{4}$ times binomial case insertion loss

• This also increases at 20N dB/decade