#### Filter Approximation Theory

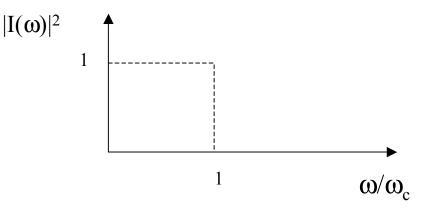
Butterworth, Chebyshev, and Elliptic Filters

#### **Approximation Polynomials**

- Every physically realizable circuit has a transfer function that is a rational polynomial in s
- We want to determine classes of rational polynomials that approximate the "Ideal" low-pass filter response (high-pass band-pass and band-stop filters can be derived from a low pass design)
- Four well known approximations are discussed here:
  - Butterworth: Steven Butterworth,"On the Theory of Filter Amplifiers", Wireless Engineer (also called Experimental Wireless and the Radio Engineer), vol. 7, 1930, pp. 536-541
  - Chebyshev: Pafnuty Lvovich Chebyshev (1821-1894) Russia
    Cyrillic alphabet Spelled many ways
    Чебышёв
  - Elliptic Function: Wilhelm Cauer (1900-1945) Germany
    U.S. patents 1,958,742 (1934), 1,989,545 (1935), 2,048,426 (1936)
  - Bessel: Friedrich Wilhelm Bessel, 1784 1846

### Definitions

• Let  $|H(\omega)|^2$  be the approximation to the ideal low-pass filter response  $|I(\omega)|^2$ 



Where  $\omega_c$  is the ideal filter cutoff frequency (it is normalized to one for convenience)

#### Definitions - 2

•  $|H(\omega)|^2$  can be written as

 $|H(\omega)|^{2} = \frac{1}{1 + \varepsilon^{2} F^{2}(\omega)}$ Where F(\omega) is the "Characteristic Function" which attempts to approximate:

$$F(\omega)$$
 1  $(\omega)$  1  $(\omega)$ 

- This cannot be done with a finite order polynomial
- $-\epsilon$  provides flexibility for the degree of error in the passband or stopband.

### Filter Specification

•  $|H(\omega)|^2$  must stay within the shaded region  $|H(\omega)|^2$ 

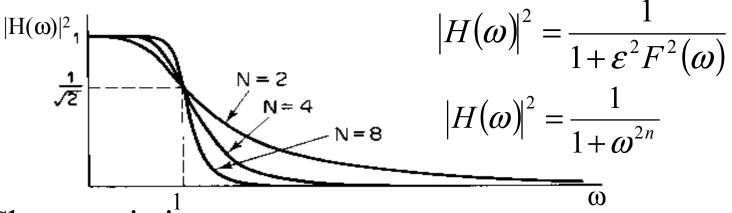
Pass band

Transition Region
 Note that this is an incomplete specification. The phase response and transient response are also important and need to be appropriate for the filter application

Stop band ω

#### Butterworth

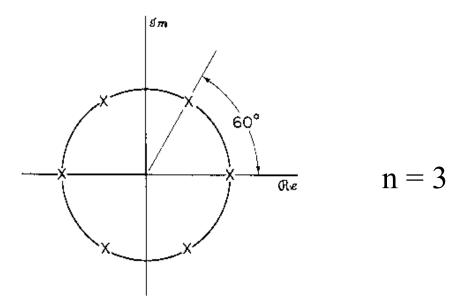
•  $F(\omega) = \omega^n$  and  $\varepsilon = 1$  and



- Characteristics
  - Smooth transfer function (no ripple)
  - Maximally flat and Linear phase (in the pass-band)
  - Slow cutoff ☺

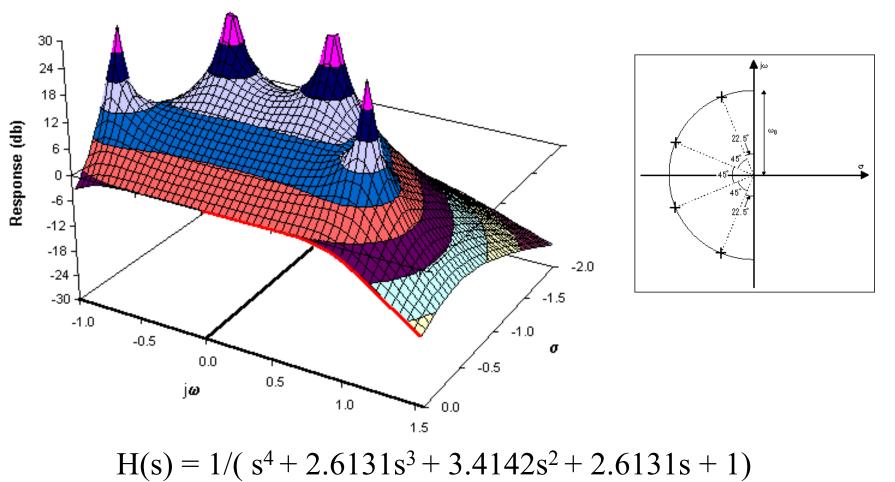
### Butterworth Continued

• Pole locations in the s-plane at:  $|H(\omega)|^2 = \frac{1}{1 + \omega^{2n}}$  $\omega^{2n} = -1 \text{ or } \omega = (-1)^{(1/2n)}$ 

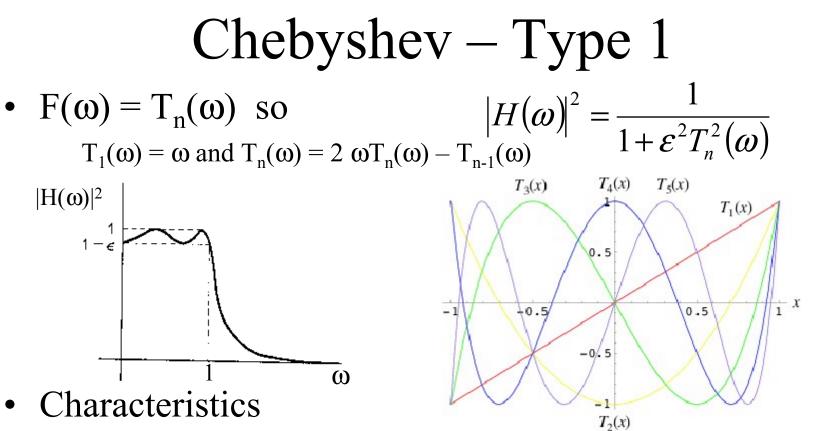


- Poles are equally spaced on the unit circle at  $\theta = k\pi/2n$ .
- H(s) only uses the n poles in the left half plane for stability.
- There are no zeros

#### Butterworth Filter |H(s)| for n=4

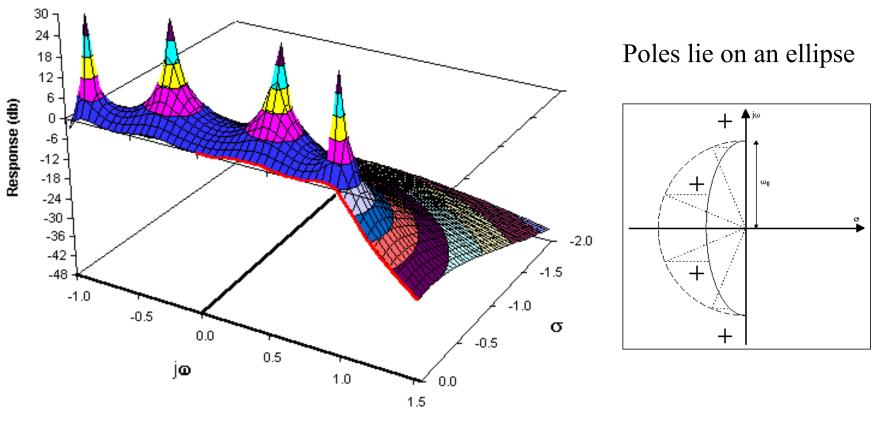


Filter Approximation Theory



- Controlled equiripple in the pass-band
- Sharper cutoff than Butterworth
- Non-linear phase (Group delay distortion)  $\otimes$
- Chebyshev type 2 moves the ripple into the stop-band

### Chebyshev |H(s)| for n=4, r=1 (Type 1)

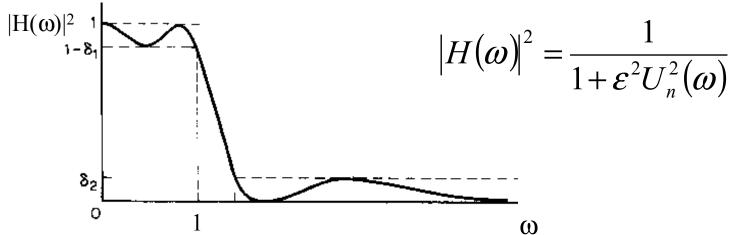


 $H(s) = 0.2457/(s^4 + 0.9528s^3 + 1.4539s^2 + 0.7426s + 0.2756)$ 

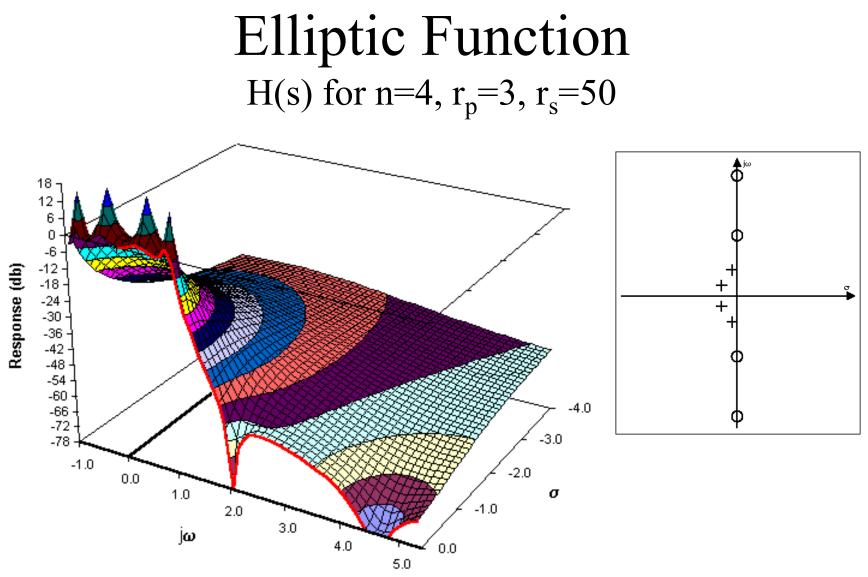
Filter Approximation Theory

# **Elliptic Function**

•  $F(\omega) = U_n(\omega)$  – the Jacobian elliptic function



- S-Plane
  - Poles approximately on an ellipse
  - Zeros on the j $\omega$ -axis
- Characteristics
  - Separately controlled equiripple in the pass-band and stop-band
  - Sharper cutoff than Chebyshev (optimal transition band)
  - Non-linear phase (Group delay distortion)  $\otimes$

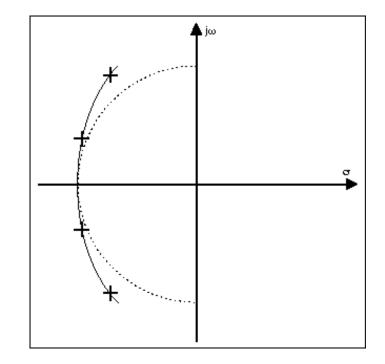


 $H(s) = (0.0032s^4 + 0.0595s^2 + 0.1554)/(s^4 + 0.5769s^3 + 1.2227s^2 + 0.4369s + 0.2195)$ 

Filter Approximation Theory

## Bessel Filter

- Butterworth and Chebyshev filters with sharp cutoffs (high order) carry a penalty that is evident from the positions of their poles in the s plane. Bringing the poles closer to the jω axis increases their Q, which degrades the filter's transient response. Overshoot or ringing at the response edges can result.
- The Bessel filter represents a trade-off in the opposite direction from the Butterworth. The Bessel's poles lie on a locus further from the  $j\omega$  axis. Transient response is improved, but at the expense of a less steep cutoff in the stop-band.



## Practical Filter Design

- Use a tool to establish a prototype design
  - MatLab is a great choice
  - See http://doctord.webhop.net/courses/Topics/Matlab/index.htm
    for a Matlab tutorial by Dr. Bouzid Aliane; Chapter 5 is on filter design.
- Check your design for ringing/overshoot.
  - If detrimental, increase the filter order and redesign to exceed the frequency response specifications
  - Move poles near the  $j\omega$ -axis to the left to reduce their Q
  - Check the resulting filter against your specifications
    - Moving poles to the left will reduce ringing/overshoot, but degrade the transition region.