

### **EEE331**

## **Analogue Electronics**

# 9th lecture:

- · passive analogue filters
  - · general filter specification
  - passive LC filters: only inductors & capacitors (difficult at low f)
  - · leapfrog design principle
  - active RC filters: R, C & op-amps (thick or hybrid thin-film technol.)
  - · switched capacitor filters: fully IC-compliant

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#### Filter transfer function: definition

definition of filter **transfer function**: T(s)=output signal /input signal =  $V_o(s)/V_i(s)$  -> get transmission for physical frequencies f by setting  $s=j\omega=2\pi jf$ .

 $T(j\omega)=|T(j\omega)|\exp[j\phi(\omega)]$  where

 $G(\omega)$ =20 log| $T(j\omega)$ | is the **gain** [dB],  $A(\omega)$ = $-G(\omega)$  the **attenuation** [dB] and  $\phi(\omega)$  is the **phase** of transmission

- ->  $T(s)=(a_ms^m+a_{m-1}s^{m-1}+\ldots+a_0)/(s^n+b_{n-1}s^{n-1}+\ldots+b_0)$  is a ratio of 2 polynomials. The degree of the denominator, n, is called the **filter order**.
- -> condition for filter circuit to be stable: **m≤n is stability criterion**The polynomials of nominator and denominator can be factorised:

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### Filter transfer function: symmetry

statement 1: zeros & poles can be real or complex, but the latter must always

occur as **conjugate pairs** to make the expression in the bracket vanish

proof: assume p is a pole, i.e. (s-p)=0. Write  $p=\sigma+j\omega$  with real  $\sigma$  and real  $\omega$ .

then:

 $p^* = \sigma - j\omega$  and  $0 = (s - p)(s - p^*)$ 

 $= S^2 - Sp - Sp^* + pp^*$ 

 $=S^2-S(\sigma+j\omega)-S(\sigma-j\omega)+(\sigma+j\omega)(\sigma-j\omega)$ 

 $= s^2 - s\sigma - js\omega - s\sigma + js\omega + \sigma^2 - js\omega + js\omega + \omega^2$ 

 $=(s-\sigma)^2+\omega^2$  implies that

 $\omega^2 = -(s - \sigma)^2$ ,

hence  $\omega = \pm j(s - \sigma)$ ,

i.e. the absolute sign of  $\omega$  is not fixed,

i.e. p\* must also be a pole.

NB: The same reasoning ca be applied to the zeros.

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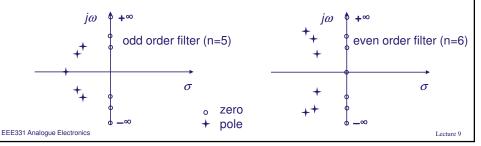


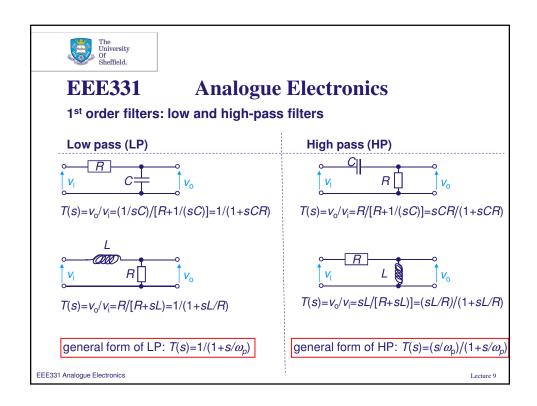
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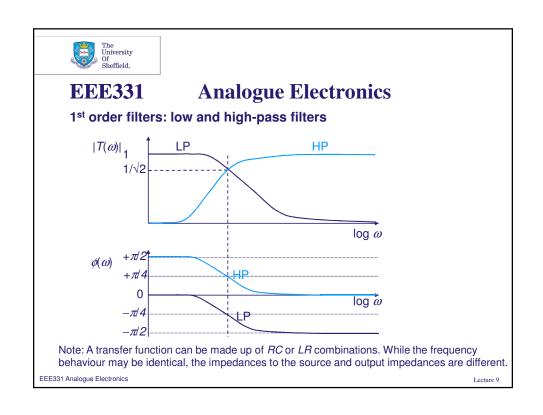
#### Filter transfer function: symmetry

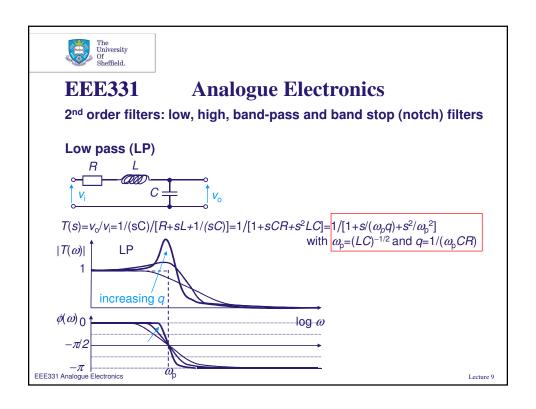
statement 2: **all poles must lie in the left half of the s-plane** for a stable filter proof: We have just seen that with  $p=\sigma+j\omega$  also  $p=\sigma-j\omega$  is a pole solution. This means complex poles always occur in conjugate pairs and we can factorise the denominator.

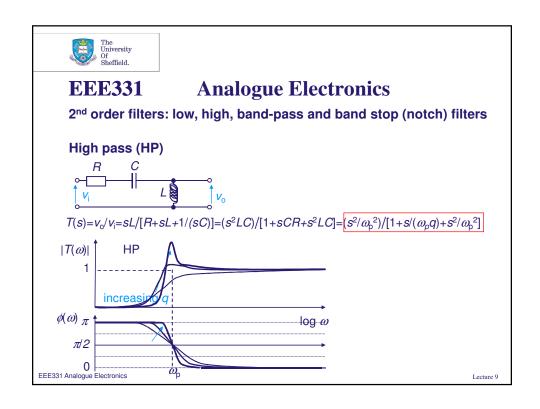
The quadratic equation  $s^2+as+b=0$  has the known solution  $s=-a/2\pm(a^2/4-b)^{1/2}$ . Hence  $s=-\sigma\pm j\omega$  is the solution to  $(s+p)(s+p^*)=s^2+2\sigma s+(\sigma^2+\omega^2)=0$ , so the real part of the solution always has the opposite sign of the terms in the brackets.

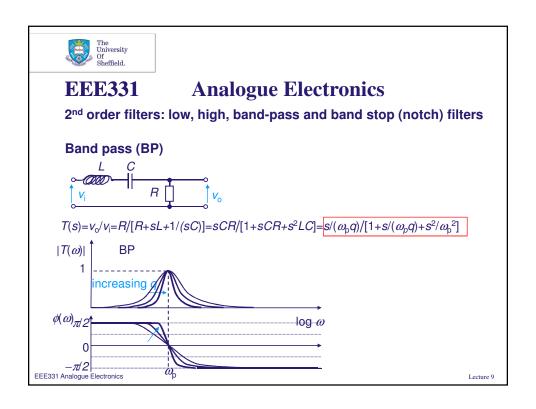


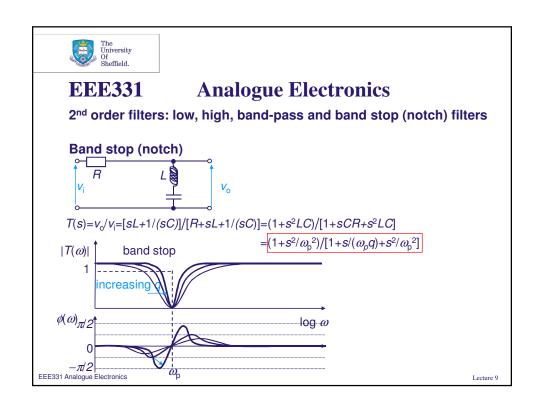














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2<sup>nd</sup> order filters: the standard from

All 2<sup>nd</sup> order filters are biquadratic transfer function of the standard form

T(s)= numerator/  $(1+s\pi q+s^2\tau^2)$ 

where the numerator decides what type of filter we have:

1 = low pass  $s\tau$  = band pass  $s^2\tau^2$  = high pass  $1+s^2\tau^2$  = band stop or notch

and the **denominator** decides the

time constant:  $\tau = (LC)^{1/2}$ ,

pole frequency:  $\omega_0 = 1/\tau = (LC)^{-1/2}$  and

quality factor:  $q=1/(\omega_0RC)=\tau/(RC)=(LC)^{1/2}/(RC)=(L/C)^{1/2}/R=\omega_0L/R$ 

It is important to get the second order transfer function into a standard form, as this allows us to define the important parameters and decide which type of filter we have. Other standard notations (for the example of the low pass filter version) are:

 $T(s) = \omega_0^2 / (\omega_0^2 + s\omega_0/q + s^2) = \omega_0^2 / (\omega_0^2 + 2\xi\omega_0 s + s^2)$ 

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2<sup>nd</sup> order filters: the meaning of the quality factor q

The **quality factor**  $q=1/(\omega_0 RC)$  of a filter has several meanings:

a) It is a measure of the magnification of the signal amplitude at resonance: Consider series connection of L, C and R. Then  $X_{\mathbb{C}}=1/(j\omega C)$  and  $X_{\mathbb{L}}=j\omega L$ .

At resonance:  $\omega = \omega_0$ . Then:  $q = 1/(\omega_0 RC) = jX_C/R = (i|X_C|)/(iR) = |v_C|/|v_R| (=|v_L|/|v_R|)$ 

b) It is the product of resonance frequency, stored energy and rate of energy dissipation:

Consider a current  $i=i_p \sin \omega_0 t$  at resonance frequency  $\omega_0=(LC)^{-1/2}$ 

The instantaneous energy stored in *L* then is:  $W_L = \frac{1}{2}Li^2 = i_p^2 \sin^2 \omega_0 t / (2\omega_0^2 C)$ 

The voltage across C is given by:  $V = Q/C = 1/C \int i dt = -1/(\omega_0 C) i_p \cos \omega_0 t$ 

The instantaneous energy stored in C is:  $W_C = \frac{1}{2}CV^2 = i_p^2 \cos^2 \omega_0 t / (2\omega_0^2 C)$ The total stored energy thus is the sum:  $W = i_p^2 / (2\omega_0^2 C) \frac{\sin^2 \omega_0 t + \cos^2 \omega_0 t}{(2\omega_0^2 C)}$ 

The rate of energy dissipation is:  $P = VI = Ri^2 = \frac{1}{2}Ri_p^2$ 

Thus  $\omega_0 W/P = q$ , i.e. q increases with the stored energy.

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 $2^{nd}$  order filters: the meaning of the quality factor q

The **quality factor**  $q=1/(\omega_0RC)$  of a filter has several meanings:

c) It measures the rate change of reactance:

Consider magnitude of reactance:  $|X| = \omega L - 1/(\omega C)$ Differenttaion yields:  $dX/d\omega = L + 1/(\omega^2 C)$ 

At resonance  $\omega=\omega_0=(LC)^{-1/2}$ :  $\mathrm{d} X/\mathrm{d} \omega |_{\mathrm{at}\;\omega=\omega_0}=1/(\omega_0^2C)+1/(\omega_0^2C)=2/(\omega_0^2C)$  Thus  $q=1/(\omega_0CR)=\omega_0/(2R)\times 2/(\omega_0^2C)=\omega_0/(2R)$  d $X/\mathrm{d} \omega |_{\mathrm{at}\;\omega=\omega_0}$  This means that q increases with the rate of change of reactance.

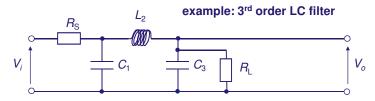
d) It measures the stability of the system:  $\lim_{a\to\infty} T(s) = \text{numerator}/(1+s\pi q + s^2\tau^2) = \text{numerator}/(1+s^2\tau^2)$  describes a simple harmonic oscillator without damping (equivalent to unity gain at  $\phi$ =180°)

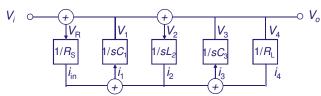
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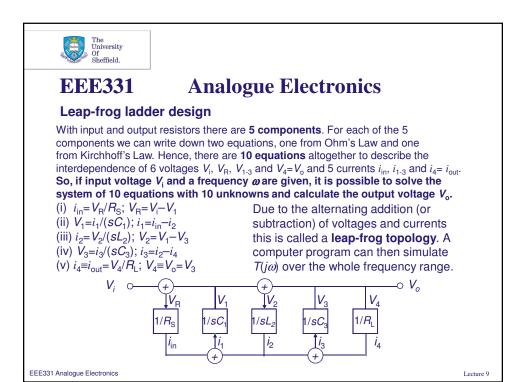
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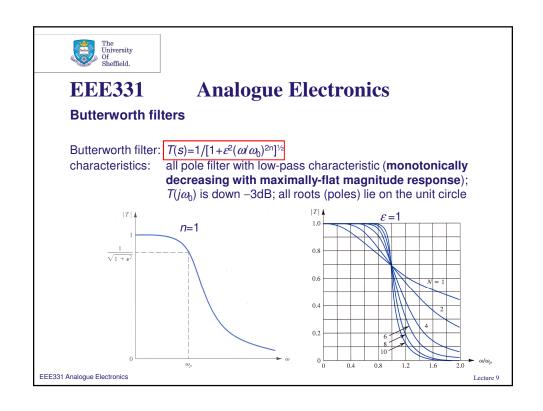
#### Leap-frog ladder design





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### **Butterworth filters**

Butterworth filter:  $T(s)=1/[1+\varepsilon^2(\omega/\omega_0)^{2n}]^{1/2}$ 

all pole filter with low-pass characteristic (monotonically characteristics:

decreasing with maximally-flat magnitude response);  $T(i\omega_{\rm h})$  is down –3dB; all roots (poles) lie on the unit circle

normalised Butterworth polynomials

```
order n denominator B_n(s) (normalised to \omega_0=1 rad/s)
          (s+1)
          (s^2+1.414s+1)
3
          (s+1)(s^2+s+1)
          (s^2+0.765s+1)(s^2+1.848s+1)
5
          (s+1)(s^2+0.618s+1)(s^2+1.618s+1)
6
          (s^2+0.518s+1) (s^2+1.414s+1) (s^2+1.932s+1)
          (s+1)(s^2+0.445s+1)(s^2+1.247s+1)(s^2+1.802s+1)
          (s^2+0.390s+1)(s^2+1.111s+1)(s^2+1.663s+1)(s^2+1.962s+1)
```

example: want a low-pass Butterworth filter with -40dB attenuation of T(s)=1/B(s) at  $\omega/\omega_0=2$ : 20 log { $[1/(1+2^{2n})]^{\frac{1}{2}}$ }=-40 gives  $2^{2n}$ = $10^4$ -1, i.e. n=6.64≈7

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#### **Chebyshev filters**

Chebyshev filter:  $T(s)=1/[1+\varepsilon^2C_n^2(\omega/\omega_0)]^{1/2}$ 

where the  $C_n$  are Chebyshev polynomials defined by  $C_n(\omega/\omega_0) = \begin{cases} \cos(n\cos^{-1}\omega/\omega_0) & \text{for } 0 \le \omega/\omega_0 \le 1\\ \cosh(n\cosh^{-1}\omega/\omega_0) & \text{for } \omega/\omega_0 > 1, \end{cases}$ 

roots (poles) lie on an ellipse

characteristics:

all pole filter with low-pass characteristic (steep but .Cutoff frequency with significant ripples); Gain= - $T(j\omega_0)$  is down -3dB; parameter  $\varepsilon$  is related to  $\,\widehat{\mathbf{g}}\,^{-20}$ passband ripple  $\gamma$ in dB by  $\varepsilon^2 = 10^{0.1} - 1$  $(\varepsilon=0.3493 \text{ for } 0.5\text{dB},$  $\varepsilon$ =0.5089 for 1dB and  $\varepsilon$ =0.7648 for 2dB ripples)  $\omega/\omega_0$ 

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