

Some Interesting Filter Design Configurations and Transformations Normally Not Found in The General Literature

Speaker: Arthur Williams – Chief Scientist Telebyte Inc.

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TOPICS

- Design of D-Element Active Low-Pass Filters and a Bidirectional Impedance Converter for Resistive Loads
- Active Adjustable Amplitude and Delay Equalizer Structures
- High-Q Notch filters
- Q-Multiplier Active Band-Pass filters
- A Family of Zero Phase-Shift Low-Pass Filters
- Some Useful Passive Filter Transformations to Improve Realizability
- Miscellaneous Circuits and "Tricks"

Frequency and Impedance Scaling from Normalized Circuit

Frequency Scaling

 $FSF = \frac{\text{desired reference frequency}}{\text{existing reference frequency}}$







Denormalized low-pass filter scaled to 1000Hz: (a) LC filter; (b) active filter; (c) frequency response.

Impedance scaling can be mathematically expressed as

$$R' = ZxR$$
$$L' = ZxL$$
$$C' = \frac{C}{7}$$

Frequency and impedance scaling are normally combined into one step rather than performed sequentially. The denormalized values are then given by

$$L' = ZxL/FSF$$

$$C' = \frac{C}{Z \times FSF}$$



Impedance-scaled filters using Z=1K : (a) LC filter; (b) active filter.

Design of D-Element Active Low-Pass Filters and a Bi-Directional Impedance Converter for Resistive Loads

Generalized Impedance Converters (GIC)

$$\mathbf{Z}_{11} = \frac{\mathbf{Z}_1 \mathbf{Z}_3 \mathbf{Z}_5}{\mathbf{Z}_2 \mathbf{Z}_4}$$

By substituting RC combinations for Z_1 through Z_5 a variety of impedances can be realized.



If Z_4 consists of a capacitor having an impedance 1/sC where s=j ω and all other elements are resistors, the driving point impedance becomes:

$$\mathbf{Z}_{11} = \frac{\mathbf{s}\mathbf{C}\mathbf{R}_1\mathbf{R}_3\mathbf{R}_5}{\mathbf{R}_2}$$

The impedance is proportional to frequency and is therefore identical to an inductor having an inductance of:



Note: If R₁ and R₂ and part of a digital potentiometer the value of L can be digitally programmable.

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D Element

If both Z_1 and Z_3 are capacitors C and Z_2, Z_4 and Z_5 are resistors, the resulting driving point impedance becomes:

$$Z_{11} = \frac{R_5}{s^2 C^2 R_2 R_4}$$
An impedance proportional
to 1/s² is called a D Element.
$$Z_{11} = \frac{1}{s^2 D}$$
where: $D = \frac{C^2 R_2 R_4}{R_5}$

If we let C=1F,R₂=R₅=1 Ω and R₄=R we get D=R so:

$$\mathbf{Z}_{11} = \frac{1}{\mathbf{s}^2 \mathbf{R}}$$

If we let s=jω the result is a Frequency Dependant Negative Resistor FDNR

$$\mathbf{Z}_{11} = \frac{1}{-\omega^2 \mathbf{R}}$$

D Element Circuit



A transfer function of a network remains unchanged if all impedances are multiplied (or divided) by the <u>same</u> factor. This factor can be a fixed number or a variable, as long as <u>every</u> impedance element that appears in the transfer function is multiplied (or divided) by the same factor.

The 1/S transformation involves multiplying all impedances in a network by 1/S.

The 1/S Transformation

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Element	Impedance	Transformed Element	Transformed Impedance
}L	sL	₹L.	L
⊥ T ^c	1 sC		$\frac{1}{s^2C}$
₹ R	R		R s

•.: •

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Design of Active Low-Pass filter with 3dB point at 400Hz using D Elements



Elliptic Function Low-Pass filter using GICs

Normalized Elliptic Function Filter Requirements: 0.5dB Maximum at 260Hz C11 20 0=75° 60dB Minimum at 270Hz N=11 R_{db}=0.18dB Ω s=1.0353 60.8dB Steepness factor=1.0385 L3 = 1.500 H L5 = 08272 H Lg = 1.093 H 1 Q L1 = 1.216 H L7 = 0.6733 H L11= 0.8788 H L10 = 0.6515 H Normalized L6=2.373 H L2 = 0.1633 H L4 = 1.145 H La = 1.881 H 210 **Elliptic Function Filter** C2 = 1.240 F C6 = 0.3914 F Ca = 0.4739 F C10 = 0.8493 F C4 = 0.6856 F Ω₆ = 1.0375 Ω₈ = 1.0592 Ω2 = 2.2222 Ω4 = 1.1286 Ω10=1.3443 (a) 1F 1.216 D 1.500 A 0.8272 Ω 0.6733 n 1.093 n 0.8788 **Ω** 0.1633 n ₹1.145 Ω 2 373 Ω 1.881 n € 0.6515 A 1/S Transformation 1F = 1.240 _____0.6856 0 3914 0.4739 0.8493 (b) 1 F 1.216 Ω 1500 A 0 8272 A 06733 n 1.093 n 08788 **Ω** 0.1633 n € 1.145 Ω \$2.373 Ω 1.881 Ω 0.6515 D 1 F 1 F 1 F 1F **Realization of D Elements** 1 Ω 1 Ω 1 Ω 1 Ω 1 0 1 F 1 F 0.4739 D 0.8493 Ω 1.240 A ≥0.6856 Ω €0.3914 Ω 1 F 1 Ω 1Ω 1 Ω 1Ω 210

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Note: 1 meg termination resistor is needed to provide DC return path.

Bi-Directional Impedance Converter for Matching D Element Filters Requiring Capacitive Loads to Resistive Terminations



Value of R Arbitrary R_s is source and load resistive terminations C_{GIC} is D Element Circuit Capacitive Terminations

Active Amplitude and Delay Equalizer Structures



T(s) is a second-order bandpass network having a gain of +1 at center frequency F_r

As a function of Q the circuit has a group delay as follows:



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- •This architecture can be used as an amplitude equalizer by varying K.
- •If K=2 the circuit can be used as a delay equalizer by varying Q.
- •By varying both K and Q this architecture can be used for both amplitude and delay equalization.

There will be an interaction between amplitude and delay as K is varied to change the amount of amplitude equalization.



Tgd,max=4R₁C R2=R3= $\frac{1}{\omega_r C}$

C,R,R' and R" can be any convenient values



Adjustable Delay and Amplitude Equalizer

By introducing a potentiometer into the circuit amplitude equalization can also be achieved The amplitude equalization at ω_r is given by:

 $A_{db} = 20 \text{ Log } (4\text{K-1})$

Where K from 0.25 to 1 covers an amplitude equalization range of $-\infty$ to +9.54 dB



To extend the amplitude equalization range beyond +9.54 dB the following circuit can be used. The amplitude equalization at ω_r given by:

 $A_{db} = 20 \text{ Log} (2K-1)$

where a K variation from 0.5 to ∞ results in an infinite range of equalization capability. In reality \pm 12dB has been found to be more than adequate so K will vary from 0.626 to 2.49.



Simplified Adjustable Amplitude Equalizer

The following circuit combines a fixed Q bandpass section with a summing amplifier to provide a low complexity adjustable amplitude equalizer. The design equations are given by:



 $A_{\rm dB} = 20 \log\left(\frac{1}{K} - 1\right)$

Where K will range from 0 to 1 for an infinite range of amplitude equalization.



To compute the desired Q first define f_b corresponding to one-half the boost(or null) desired in dB.

Then:



High-Q Notch Filters



This circuit is in the form of a bridge where a signal is applied across terminals' 1 and 2 the output is measured across terminals' 3 and 4. At ω =1 all branches have equal impedances of 0.707 \angle -45°so a null occurs across the output.



The circuit is redrawn in figure B in the form of a lattice. Circuit C is the Identical circuit shown as two lattices in parallel.



There is a theorem which states that any branch in series with both the Z_A and Z_B branches of a lattice can be extracted and placed outside the lattice. The branch is replaced by a short. This is shown in figure D above. The resulting circuit is known as a <u>Twin-T</u>. This circuit has a null at 1 radian for the normalized values shown.



To calculate values for this circuit pick a convenient value for C. Then

$$R_1 = \frac{1}{2\pi f_0 C}$$

The Twin-T has a Q (f_0/BW_{3dB}) of only $\frac{1}{4}$ which is far from selective.



Circuit A above illustrates bootstrapping a network β with a factor K. If β is a twin-T the resulting Q becomes:

$$Q = \frac{1}{4(1-K)}$$

If we select a positive K <1, and sufficiently close to 1, the circuit Q can be dramatically increased. The resulting circuit is shown in figure B.

Bridged-T Null Network



Impedance of a center-tapped parallel resonant circuit <u>at resonance</u> is $\omega_r LQ$ total and $\omega_r LQ/4$ from end to center tap (due to N² relationship). Hence a phantom negative resistor of $-\omega_r LQ/4$ appears in the equivalent circuit which can be cancelled by a positive resistor of $\omega_r LQ/4$ resulting in a very deep null at resonance (60dB or more).

Adjustable Q and Frequency Null Network



T(s) can be any bandpass circuit having properties of unity gain at f_r , adjustable Q and adjustable f_r .

Q Multiplier Active Bandpass Filters



The middle term of the denominator has been modified so the circuit Q is given by $Q/(1-\beta)$ where $0<\beta <1$. The Q can then be increased by the factor $1/(1-\beta)$. Note that the circuit gain is increased by the same factor.

A simple implementation of this circuit is shown in figure B. The design equations are:

First calculate
$$\beta$$
 from $\beta = 1 - \frac{Q_r}{Q_{eff}}$

where Q_{eff} is the overall circuit Q and Q_r is the design Q of the bandpass section.

The component values can be computed from:

$$R_{3} = \frac{R}{\beta}$$

$$R_{2} = \frac{Q_{r}}{\pi f_{r}C}$$

$$R_{1b} = \frac{R_{1a}}{2Q_{r}^{2}-1}$$

$$R_{4} = R$$

$$R_{1a} = \frac{R_{2}}{2}$$

$$R_{5} = \frac{R}{(1-\beta)A_{r}}$$

Where R and C can be conveniently chosen.



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A Family of Zero Phase Shift Low-Pass Filters



note near-zero phase slope for high ripples



Transformation to Band-Pass Filter



Some Useful Passive Filter Transformations to Improve Realizability



Increases value of L

Narrow Band Approximations



This transformation can be used to reduce the value of a terminating resistor and yet maintain the narrow-band response.

and

$$C_2 = \frac{1}{\omega_0 \sqrt{R_1 R_2 - R_2^2}}$$
$$C_1 = C_T - \frac{1}{\omega_0} \sqrt{\frac{R_1 - R_2}{R_1^2 R_2}}$$

where the restrictions $R_2 < R_1$ and $(R_1 - R_2)/(R_1^2 R_2) < \omega_0^2 C_T^2$ apply.



The following example illustrates how this approximation can reduce the source impedance of a filter.



Using the Tapped Inductor



An inductor can be used as an auto-transformer by adding a tap



Resonant circuit capacitor values can be reduced





Effect of leakage inductance can be minimized by splitting capacitors which adds additional poles

Exponentially Tapered Network



Method of exponentially tapering a network to a higher (or lower) load impedance value with minimal effect on response



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Bandpass output 44

AII-PASS DELAY LINE SECTION



SINGLE TRANSFORMER HYBRID

