

# 1 Definitions

$\omega_c$ : Cutoff Frequency

$\omega_{c,h}$ : High Cutoff Frequency (Bandpass)

$\omega_{c,l}$ : Low Cutoff Frequency (Bandpass)

$\omega_0$ : Center Frequency

$B_p$ :  $\omega_{c,h} - \omega_{c,l}$

$sbd_{dB}$ : Stopband Deviation in Decibels

$pbd_{dB}$ : Passband Deviation in Decibels

$G_0$ : Passband Gain

$F(s)$ : Filter Equation

$n$ : Filter Order

$k_f$ : Scaling Factor for the Cutoff Frequency

$\Delta$ : Difference / Derivative

$x^*$ : Complex Conjugate of  $x$

## 1.1 Group Delay

$$\frac{\Delta p}{\Delta f}$$

Where  $-180 \leq p \leq 180$  degrees and  $f$  in Hertz

# 2 Analog Filters

	Bessel	
	Butterworth	
Narrower Transition ↓	Chebyshev	↑ More Linear Passband Phase
	Elliptic	

## 2.1 Bessel

Choose  $n$  then create as follows:

$$F(s) = \frac{\theta_n(0)}{\theta_n(s)}$$

$$\theta_n(x) = \sum_{k=0}^n \frac{(2n-k)!}{2^{n-k} k! (n-k)!} x^k$$

## 2.2 Butterworth

Choose  $n$  and  $G_0$  then create as follows:

$$F(s) = \frac{G_0}{B_n(\frac{s}{w_c})}$$

For even  $n$ :

$$B_n(x) = \prod_{k=1}^{n/2} x^2 - 2\cos(\frac{\pi(2k+n-1)}{2n})x + 1$$

For odd  $n$ :

$$B_n(x) = (x+1) \prod_{k=1}^{(n-1)/2} x^2 - 2\cos(\frac{\pi(2k+n-1)}{2n})x + 1$$

### 2.2.1 Chebyshev

Choose  $G_0$ ,  $n$ ,  $\omega_c$ , and  $pb_{dB}$  then create as follows:

$$F(s) = \frac{G_0 X_n(s)}{V_n(s)}$$

$$\epsilon = \sqrt{10^{\frac{-pb_{dB}}{10}} - 1}$$

$$\alpha = \frac{1}{\epsilon} + \sqrt{1 + \epsilon^{-2}}$$

$$A = \frac{\alpha^{\frac{1}{n}} - \alpha^{\frac{-1}{n}}}{2}$$

$$B = \frac{\alpha^{\frac{1}{n}} + \alpha^{\frac{-1}{n}}}{2}$$

For even  $n$ :

$$p_k = \omega_c(A\cos(\frac{(2k-1)\pi}{2n}) + jB\sin(\frac{(2k-1)\pi}{2n}))$$

$$V_n(s) = \prod_{k=1}^{n/2} s^2 + s(p_k + p_k^*) + p_k p_k^*$$

$$X_n(s) = 10^{-pb_{dB}/20} \prod_{k=1}^{n/2} p_k p_k^*$$

For odd  $n$ :

$$p_k = \omega_c(A\cos(\frac{k\pi}{n}) + jB\sin(\frac{k\pi}{n}))$$

$$V_n(s) = (s + A\omega_c) \prod_{k=1}^{(n-1)/2} s^2 + s(p_k + p_k^*) + p_k p_k^*$$

$$X_n(s) = A\omega_c \prod_{k=1}^{(n-1)/2} p_k p_k^*$$

Side Note:

Chebyshev Polynomials:

$$T_n(x) = \begin{cases} T_0(x) = 1 \\ T_1(x) = x \\ T_{n+1}(x) = 2xT_n(x) - T_{n-1}(x) \end{cases} = \begin{cases} \cos(ncos^{-1}(x)), |x| \leq 1 \\ \cosh(ncosh^{-1}(x)), x \geq 1 \\ -1^n \cosh(ncosh^{-1}(-x)), x \leq -1 \end{cases}$$

$$|F(s)| = \sqrt{\frac{1}{1 + \epsilon^2 T_n^2(\frac{\omega}{\omega_c})}}$$

### 2.2.2 Elliptic / Cauer

### 2.2.3 Inverse Chebyshev / Type II Chebyshev

Choose  $G_0$ ,  $n$ ,  $\omega_c$ , and  $sbd_{dB}$  then create as follows:

$$F(s) = \frac{G_0 X_n(s)}{V_n(s)}$$

$$A_s = 10^{\frac{sbd_{dB}}{10}}$$

$$\epsilon = \sqrt{\frac{A_s}{1 - A_s}}$$

$$u = \frac{\sinh^{-1}(\frac{1}{\epsilon})}{n}$$

$$z_k = j\omega_c \sec\left(\frac{(2k-1)\pi}{2n}\right)$$

$$p_k = \frac{\omega_c}{\sinh(u) \sin\left(\frac{(2k-1)\pi}{2n}\right) + j \cosh(u) \cos\left(\frac{(2k-1)\pi}{2n}\right)}$$

For even n:

$$V_n(s) = \prod_{k=1}^{k=n/2} s^2 + s(p_k p_k^*) + p_k p_k^*$$

$$X_n(s) = \prod_{k=1}^{k=n/2} \frac{p_k p_k^*}{z_k z_k^*} (s^2 + s(z_k + z_k^*) + z_k z_k^*)$$

For odd n:

$$V_n(s) = (s + p_n) \prod_{k=1}^{k=(n-1)/2} (s^2 + s(p_k p_k^*) + p_k p_k^*)$$

$$X_n(s) = \sqrt{p_n p_n^*} \prod_{k=1}^{k=(n-1)/2} \frac{p_k p_k^*}{z_k z_k^*} (s^2 + s(z_k + z_k^*) + z_k z_k^*)$$

## 2.3 Filter Transformations

### 2.3.1 Frequency Scaling

Replace  $s$  with  $\frac{s}{k_f}$

### 2.3.2 Lowpass to Highpass

Replace  $s$  with  $\frac{1}{s}$

### 2.3.3 Lowpass to Bandpass

Replace  $s$  with  $\frac{s^2 + \omega_0^2}{B_p s}$

### 2.3.4 Lowpass to Bandstop

Replace  $s$  with  $\frac{B_p s}{s^2 + \omega_0^2}$

### 2.3.5 Lowpass to Notch

Notch is synonymous with ‘Bandstop’ and ‘Band-reject’

Replace  $s$  with  $\frac{B_p s}{s^2 + \omega_0^2}$

## 3 Digital Filters

### 3.1 Bilinear Transformation

Transforms any analog filter into a IIR digital filter. Phase response will differ from the analog version.

Form desired analog filter then replace  $s$  as follows:  $s = \frac{2}{T} \frac{1-z^{-1}}{1+z^{-1}}$

where  $T$  is the sample rate and  $z^{-1}$  is the previous data point

## 3.2 Finite Impulse Response (FIR)

No feedback and linear phase

### 3.2.1 Window Method

$$F(n) = [Window] \times [Impulse Type]$$

Window	Peak Sidelobe	Main Lobe Width	Peak Approximation Error
Rectangular	0	0	0
Hanning	0	0	0
Hamming	0	0	0
Blackmann	0	0	0

Window	Equation
Rectangular	0
Hanning	$\sin^2(\frac{n\pi}{N})$
Hamming	$0.54 - 0.46\cos(\frac{2\pi n}{N})$
Blackmann	$0.42 - 0.5\cos(\frac{2\pi n}{N}) + 0.08\cos(\frac{4\pi n}{N})$

Impulse Type	Equation
Low Pass	$\frac{\omega_c}{\pi} \text{sinc}(\frac{n\omega_c}{\pi})$
High Pass	$\text{sinc}(n) - \frac{\omega_c}{\pi} \text{sinc}(\frac{n\omega_c}{\pi})$
Band Pass	$\frac{\omega_{c,h}}{\pi} \text{sinc}(\frac{n\omega_{c,h}}{\pi}) - \frac{\omega_{c,l}}{\pi} \text{sinc}(\frac{n\omega_{c,l}}{\pi})$

## 4 Implementations

### 4.1 Passive

#### 4.1.1 Pi Filter

#### 4.1.2 T Filter

### 4.2 Active

#### 4.2.1 Sallen-Key

$$\frac{V_{OUT}}{V_{IN}} = \frac{Z_3 Z_4}{Z_1 Z_2 + Z_3(Z_1 + Z_2) + Z_3 Z_4}$$

For Lowpass:  $Z_1, Z_2$  become resistors and  $Z_3, Z_4$  become capacitors

For Highpass:  $Z_3, Z_4$  become resistors and  $Z_1, Z_2$  become capacitors

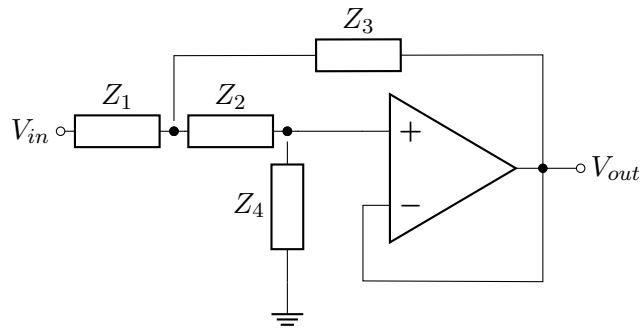


Figure 1: Sallen-Key Generic

#### 4.2.2 Phase Lead-Lag Compensator