SigmaStudio Filter Coefficient Calculations

This document shows how to calculate B0, B1, B2, A1, and A2 for various IIR filter types in SigmaStudio. **Sin** = sine, **Cos** = cosine, **Tan** = tangent, **Sinh** = hyperbolic sine, **Log** = base 10 logarithm

Parametric EQ Calculation

```
Values from Control:
```

```
boost (boost of filter)
frequency (center frequency)
Q (Q of filter; must be greater than or equal to 0.01)
gain (linear gain applied to the signal)
Fs (sample rate of project)
      In the case that boost = 0,
        gainlinear = 10^{(gain / 20)}
        B0 = gainlinear
        B1 = 0
        B2 = 0
        A1 = 0
        A2 = 0
      If boost is not 0,
        Create 5 new variables: a0, omega, sn, cs, alpha, Ax
        Ax = 10^{(boost / 40)}
        omega = 2 * PI * frequency / Fs
        sn = Sin(omega)
        cs = Cos(omega)
        alpha = sn / (2 * (Q))
        a0 = 1 + (alpha / Ax)
        A1 = -(2 * cs) / a0
        A2 = (1 - (alpha / Ax)) / a0
        gainlinear = 10^{(gain / 20)} / a0
        B0 = (1 + (alpha * Ax)) * gainlinear
        B1 = -(2 * cs) * gainlinear
        B2 = (1 - (alpha * Ax)) * gainlinear
```

Tone Control

Values from Control:

```
Freq_T (treble cutoff frequency)
Boost_T (treble boost)
Freq_B (bass cutoff frequency)
Boost_B (bass boost)
Fs (sample rate of project)
                     Boost_T = 10^{(Boost_T / 20)}
                     Boost_B = 10^{(Boost_B / 20)}
                     A T = Tan(PI * Freq T / Fs)
                     A_B = Tan(PI * Freq_B / Fs)
                     Knum_T = 2 / (1 + (1 / Boost_T))
                     Kden_T = 2 / (1 + Boost_T)
                     Knum_B = 2 / (1 + (1 / Boost_B))
                     Kden_B = 2 / (1 + Boost_B)
                     a10 = A_T + Kden_T
                                                         b10 = A_T + Knum_T
                     a11 = A_T - Kden_T
                                                         b11 = A_T - Knum_T
                     a20 = (A B * Kden B) + 1
                                                         b20 = (A_B * Knum_B) - 1
                     a21 = (A_B * Kden_B) - 1
                                                         b21 = (A_B * Knum_B) + 1
                     a0 = a10 * a20
                     A1 = ((a10 * a21) + (a11 * a20)) / a0
                                                                  / a0
                     A2 = a11 * a21
                     gainlinear = 10^{(cell\_gain / 20)}
                     B0 = (b10 * b20) / a0 * gainlinear
                     B1 = ((b10 * b21) + (b11 * b20)) / a0 * gainlinear
                     B2 = (b11 * b21) / a0 * gainlinear
```

For double 1st Order Filter cells, simply use 2 Cascaded 1st Order Filters

All Pass

```
Values from control:
frequency (center frequency)
Q (Q of filter)
gain1 (linear gain)
fs (sample rate of project)
                     gain1 = 10^{(gain / 20)}
                     omega = 2 * PI * frequency / fs
                     sins = Sin(omega)
                     coss = Cos(omega)
                     alpha = sins / (2 * Q)
                     norm = 1 + alpha
                     B0 = gain1 * (1 - alpha) / norm
                     B1 = gain1 * ( - 2 * coss) / norm
                     B2 = gain1 * (1 + alpha) / norm
                     A1 = -2 * coss / norm
                     A2 = (1 - alpha) / norm
```

Notch Filter

Chebyshev Low Pass

```
Values from control:
frequency (cutoff frequency; must be 1 Hz or greater)
gain (linear gain)
ripple (ripple of filter; must be 0.1 or greater)
Fs (sample rate of project)
       wp = (2 * PI * frequency) / Fs
       Omegap = Tan(wp / 2)
epass = (10^{(0.1 * ripple)} - 1)^{0.5}
       alpha = (0.5) * Log(1 / epass + (1 / (epass^2 + 1))^{0.5}
       Omega0 = Omegap * Sinh(alpha);
       theta = (PI / 4) * 3
       Omega1 = Omegap * Sin(theta)
       H0 = (1 / (1 + epass^2))^{0.5}
       Den = 1 - (2 * Omega0 * Cos(theta)) + Omega0^2 + Omega1^2
G = (Omega0^2 + Omega1^2) / Den
       A1 = (2 * (Omega0^2 + Omega1^2 - 1)) / Den
       A2 = (1 + (2 * Omega0 * Cos(theta)) + Omega0^2 + Omega1^2) / Den B0 = H0 * G * 10^{(gain / 20)}
       B1 = B0 * 2
       B2 = B0
```

ChebyshevHighPass

```
Values from control:
frequency (cutoff frequency; must be 1 Hz or greater)
gain (linear gain)
ripple (ripple of filter; must be 0.1 or greater)
Fs (sample rate of project)
       wp = (2 * PI * frequency) / Fs
       Omegap = 1 / Tan(wp / 2)
epass = (10^{(0.1 * ripple)} - 1)^{0.5}
       alpha = (0.5) * Log(1 / epass + (1 / (epass<sup>2</sup>) + 1))<sup>0.5</sup>
       Omega0 = Omegap * Sinh(alpha)
       theta = (PI / 4) * 3
       Omega1 = Omegap * Sin(theta)
       H0 = (1 / (1 + (epass^2)))^{0.5}
       Den = 1 - 2 * Omega0 * Cos(theta) + (Omega0^2) + (Omega1^2)
       G = ((Omega0^2) + (Omega1^2)) / Den
       A1 = (-2 * ((Omega0^2) + (Omega1^2) - 1)) / Den
       A2 = (1 + 2 * Omega0 * Cos(theta) + (Omega0^2) + (Omega1^2)) / Den B0 = H0 * G * <math>(10^{(gain / 20)})
       B1 = - B0 * 2
       B2 = B0
```

Linkwitz-Riley - 12 dB/oct = 2 cascaded 1st order butterworths (2 biquads)

```
frequency (the cutoff frequency)
First Order type (the type of filter, can be lowpass, highpass, or allpass)
g (the linear gain)
fs (the sample rate of the project)
       1<sup>st</sup> Order Butterworth
      gain = (10^{(g / 20)})
      omega, sn, cs, alpha, a0
      omega = 2 * PI * frequency / fs
      sn = Sin(omega)
      cs = Cos(omega)
      a0 = sn + cs + 1;
      For a First Order lowpass...
        A1 = (sn - cs - 1) / a0
        B0 = gain * sn / a0
        B1 = gain * sn / a0
      For a First Order highpass...
        A1 = (sn - cs - 1) / a0
        B0 = gain * (1 + cs) / a0
        B1 = - gain * (1 + cs) / a0
      For a First Order allpass...
        A1 = (2.7^{(-2*PI*frequency/fs)})
        B0 = - gain * A1
        B1 = gain
```

Linkwitz-Riley - 24 dB/oct = 2 cascaded 2nd order butterworths (2 biquads)

```
frequency (the cutoff frequency)
gain (the linear gain)
Fs (the sample rate of the project)
                      2<sup>nd</sup> Order LOWPASS
                      omega, sn, cs, alpha, a0;
                      omega = 2 * PI * frequency / Fs
                      sn = Sin(omega)
                      cs = Cos(omega)
                      alpha = sn / (2 * (1 / (2)^{0.5}))
                      a0 = 1 + alpha
                      A1 = -(2 * cs) / a0
                      A2 = (1 - alpha) / a0
                      B1 = (1 - cs) / a0 * (10^{(gain / 20)})
                      B0 = B1 / 2
                      B2 = B0
                      2<sup>nd</sup> Order HIGHPASS
                      omega = 2 * PI * frequency / Fs
                      sn = Sin(omega)
                      cs = Cos(omega)
                      alpha = sn / (2 * (1 / (2)^{0.5}))
                      a0 = 1 + alpha
                      A1 = -(2 * cs) / a0
                      A2 = (1 - alpha) / a0
                      B1 = -( 1 + cs) / a0 * (10^{(gain / 20)})
                      B0 = - B1 / 2
```

B2 = B0

```
Linkwitz-Riley - 36 dB/oct = 2 cascaded 3rd order butterworths
3rd order butterworth is implemented by cascading a "HigherOrder" + 1st order
1<sup>st</sup> Filter: [HigherOrder] orderindex = 3, i = 0
2<sup>nd</sup> Filter: 1st Order Butterworth
3<sup>rd</sup> Filter: [HigherOrder] orderindex = 3, i = 0
4<sup>th</sup> Filter: 1st order Butterworth
frequency (the cutoff frequency)
gain (the linear gain)
Fs (the sample rate of the project)
orderindex (described above... changes based on type)
i (described above)
       LOW PASS HIGHER ORDER
      omega, sn, cs, alpha, a0 , orderangle
      omega = 2 * PI * frequency / Fs
      sn = Sin(omega)
      cs = Cos(omega)
         orderangle = (PI / orderindex) * (i + 0.5)
      alpha = sn / (2 * (1 / (2 * Sin(orderangle))))
      a0 = 1 + alpha
     A1 = -(2 * cs) / a0
     A2 = (1 - alpha) / a0
     B1 = (1 - cs) / a0 * (10^{(gain / 20)})
     B0 = B1 / 2
      B2 = B0
      HIGH PASS HIGHER ORDER
             omega = 2 * PI * frequency / Fs
             sn = Sin(omega)
             cs = Cos(omega)
             orderangle = (PI / orderindex) * (i + 0.5)
      alpha = sn / (2 * (1 / (2 * Sin(orderangle))))
             a0 = 1 + alpha
             A1 = -(2 * cs) / a0
             A2 = (1 - alpha) / a0
             B1 = -( 1 + cs) / a0 * (10^{(gain / 20)})
             B0 = - B1 / 2
             B2 = B0
```

Linkwitz-Riley - 48 dB/oct = 2 cascaded 4th order butterworths
4th order butterworth is implemented by cascading 2 2nd "Higher Order" using equations shown above.

```
1<sup>st</sup> Filter: orderindex = 4, i = 0

2<sup>nd</sup> Filter: orderindex = 4, i = 1

3<sup>rd</sup> Filter: orderindex = 4, i = 0

4<sup>th</sup> Filter: orderindex = 4, i = 1
```

Butterworth 12 dB/oct = One 2nd order butterworth

Butterworth 18 dB/oct = One Higher order butterworth + One 1st Order

Filt 1: [HigherOrder]: orderindex = 3, i = 0

Filt 2: 1st Order butterworth

Butterworth 24 dB/oct = 2 Higher order butterworths

Filt 1: orderindex = 4, i = 0 Filt 2: orderindex = 4, i = 1

Bessel 12 dB/oct = One 2nd order Bessel

frequency,
gain,
Fs

Low Pass Bessel

```
omega, sn, cs, alpha, a0

omega = 2 * PI * frequency / Fs

sn = Sin(omega)

cs = Cos(omega)

alpha = sn / (2 * (1 / (3)<sup>0.5</sup>))

a0 = 1 + alpha

A1 = -(2 * cs) / a0

A2 = (1 - alpha) / a0

B1 = (1 - cs) / a0 * (10<sup>(gain / 20)</sup>)

B0 = B1 / 2

B2 = B0
```

High Pass Bessel

```
omega = 2 * PI * frequency / Fs

sn = Sin(omega)

cs = Cos(omega)

alpha = sn / (2 * (1 / (3)<sup>0.5</sup>))

a0 = 1 + alpha

A1 = -( 2 * cs) / a0

A2 = (1 - alpha) / a0

B1 = -( 1 + cs) / a0 * (10<sup>(gain / 20)</sup>)

B0 = - B1 / 2

B2 = B0
```

Bessel 18 dB/oct = one 2nd order Bessel + One 1st order Butterworth

Bessel 24 dB/oct = Two 2nd order Besels