# Discretization of a Fourth-Order Butterworth Filter

This is an example on how to design a filter in the analog domain, and then use the bilinear transform to transform it to the digital domain, while preserving the cut-off frequency.

We'll be using formulas derived on the Bilinear Transform and Butterworth Filters pages

### Design criteria

In this example, we'll design a digital fourth order Butterworth low-pass filter, with a sample frequency of 360 Hz and a cut-off frequency of 45 Hz.

## Frequency Pre-Warping

As discussed in the page on the Bilinear Transform, we have to apply pre-warping to the cut-off frequency before designing a filter. If we don't the cut-off frequency will shift to an incorrect frequency when we discretize the filter.

First, let's calculate the normalized digital frequency  $\omega_{cd}$ , using the cut-off frequency  $f_c$  and the sample frequency  $f_s$ :

$$egin{aligned} f_c &= 45 \; \mathrm{Hz} \\ f_s &= 360 \; \mathrm{Hz} \end{aligned}$$
 $T_s &= rac{1}{f_s}$ 
 $\omega_c &= 2\pi f_c$ 
 $\approx 282.7 \; \mathrm{rad} \; s^{-1}$ 
 $\omega_{c,d} &= rac{\omega_c}{f_s}$ 
 $&= rac{\pi}{4} rac{\mathrm{rad}}{\mathrm{sample}} pprox 0.7854 rac{\mathrm{rad}}{\mathrm{sample}}$ 

The Nyquist-Shannon sampling theorem tells us that we can never sample frequencies higher than  $f_s/2$  without losing information. This also means that the cut-off frequency can never be higher than half of the sample frequency. Or in other words, all normalized frequencies will be in the interval  $[0, \pi]$ .

Next, we'll use the pre-warping formula we derived in the page on the Bilinear Transform, in order to calculate the analog design frequency  $\omega_{c,a}$ 

$$egin{align} \omega_{c,a} &= rac{2}{T_s} anigg(rac{\omega_{c,d}}{2}igg) \ &= 720 anigg(rac{\pi}{8}igg) pprox 298.2 ext{ rad } s^{-1} \end{split}$$

Note that this frequency is relatively close to  $\omega_c$ , but it is not the same. The higher the cut-off frequency (relative to the sample frequency), the larger the error between  $\omega_c$  and  $\omega_{c,a}$ 

#### Designing the Butterworth filter in the Analog Domain

Now that we know the pre-warped analog cut-off frequency, we can start designing the analog filter. We'll use the formula for the Butterworth low-pass filter derived in the page on Butterworth Filters:

$$H_4(s') = \frac{1}{B_4(s')} \quad \text{where } s' \triangleq \frac{s}{\omega_{c,a}}$$

$$B_4(s') = \left(s'^2 - 2\cos\left(2\pi\frac{4+1}{4\cdot 4}\right)s' + 1\right) \left(s'^2 - 2\cos\left(2\pi\frac{2+4+1}{4\cdot 4}\right)s' + 1\right)$$
(1)

Defining these constants will make the calculations much easier:

$$\alpha \triangleq -2\cos\left(\frac{5\pi}{8}\right) \\ = \sqrt{2-\sqrt{2}}$$
 (2)

$$\beta \triangleq -2\cos\left(\frac{7\pi}{8}\right) \\ = \sqrt{2 + \sqrt{2}} \tag{3}$$

$$B_4(s') = (s'^2 + \alpha s' + 1) (s'^2 + \beta s' + 1)$$
  
=  $s'^4 + s'^3(\alpha + \beta) + s'^2(\alpha \beta + 2) + s'(\alpha + \beta) + 1$  (4)

## Discretizing the Analog Filter

We can now just apply the Bilinear Transform to the analog transfer function, by substituting  $s=\frac{2}{T_s}\frac{z-1}{z+1}$ . Therefore:

$$s'=rac{2f_s}{\omega_{a.c}}rac{z-1}{z+1}$$

Again, we'll introduce a constant to simplify the expression:

$$\gamma \triangleq \frac{2f_s}{\omega_{a,c}} = \frac{2f_s}{2f_s \tan\left(\frac{\omega_{c,d}}{2}\right)} = \cot\left(\pi \frac{f_c}{f_s}\right) \\
= \cot\left(\frac{\pi}{8}\right) = 1 + \sqrt{2} \tag{5}$$

$$s' = \gamma \frac{z - 1}{z + 1} \tag{6}$$

What follows is just rearranging the expression of  $B_4(s')$  from Equation 4, using the substitution of Equation 6 Finally, we end up with an expression for the transfer function, using Equation 1, and we can determine the coefficients using the constants defined in Equations 2, 3 & 5.

$$\begin{split} B_4(s') &= s^4 + s^2(\alpha + \beta) + s^2(\alpha\beta + 2) + s'(\alpha + \beta) + 1 \\ B_4(z) &= \gamma^4 \frac{(z-1)^4}{(z+1)^4} \\ &+ \gamma^3 \frac{(z-1)^3(z+1)}{(z+1)^4} (\alpha + \beta) \\ &+ \gamma^2 \frac{(z-1)^2(z+1)^2}{(z+1)^4} (\alpha\beta + 2) \\ &+ \gamma \frac{(z-1)^2(z+1)^3}{(z+1)^4} (\alpha+\beta) \\ &+ \frac{(z+1)^4}{(z+1)^4} \end{split}$$

$$&= \frac{1}{(z+1)^4} \begin{bmatrix} \gamma^4(z-1)^4 \\ + \gamma^3(z-1)^3(z+1)(\alpha+\beta) \\ + \gamma^4(z-1)^2(z+1)^2(\alpha\beta + 2) \\ + \gamma(z-1)(z+1)^3(\alpha+\beta) \end{bmatrix}$$

$$&= \frac{1}{(z+1)^4} \begin{bmatrix} \gamma^4 & (z^4 - 4z^3 + 6z^2 - 4z + 1) \\ + \gamma^3 & (z^4 - 2z^3 - 2z^2 + 1) \end{pmatrix} (\alpha\beta + 2) \\ + \gamma & (z^4 + 2z^3 - 2z^2 + 1) \end{pmatrix} (\alpha\beta + 2) \\ + \gamma & (z^4 + 2z^3 - 2z^2 + 1) \end{pmatrix} (\alpha\beta + 2) \end{bmatrix}$$

$$&= \frac{1}{(z+1)^4} \begin{bmatrix} \gamma^4 + \gamma^3(\alpha+\beta) + \gamma^2(\alpha\beta + 2) + \gamma(\alpha+\beta) + 1) & z^4 \\ + (-4\gamma^4 - 2\gamma^3(\alpha+\beta) + 2\gamma(\alpha+\beta) + 1) & z^4 \\ + (-4\gamma^4 - 2\gamma^3(\alpha+\beta) + 2\gamma(\alpha+\beta) + 1) & z^4 \\ + (-4\gamma^4 - 2\gamma^3(\alpha+\beta) + \gamma^2(\alpha\beta + 2) - \gamma(\alpha+\beta) + 1) \end{bmatrix}$$

$$&= \frac{1}{B_4(z)} \begin{bmatrix} (\gamma^4 + \gamma^3(\alpha+\beta) + \gamma^2(\alpha\beta + 2) + \gamma(\alpha+\beta) + 1) & z^4 \\ + (-4\gamma^4 - 2\gamma^3(\alpha+\beta) + \gamma^2(\alpha\beta + 2) - \gamma(\alpha+\beta) + 1) & z^4 \\ + (-4\gamma^4 - 2\gamma^3(\alpha+\beta) + \gamma^2(\alpha\beta + 2) - \gamma(\alpha+\beta) + 1 \end{bmatrix}$$

$$&= \frac{b_4z^4 + b_3z^3 + b_2z^2 + b_1z + b_0}{a_4z^4 + a_3z^3 + a_2z^2 + a_1z + a_0}$$

$$&= \alpha_0 = \gamma^4 - \gamma^3(\alpha+\beta) + \gamma^2(\alpha\beta + 2) - \gamma(\alpha+\beta) + 1 \approx 11.79$$

$$&= \alpha_1 = -4\gamma^4 + 2\gamma^3(\alpha+\beta) + \gamma^2(\alpha\beta + 2) - \gamma(\alpha+\beta) + 1 \approx 11.79$$

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$$&= \alpha_1 = -4\gamma^4 - 2\gamma^3(\alpha+\beta) + \gamma^2(\alpha\beta + 2) + \gamma(\alpha+\beta) + 1 \approx 97.95$$

$$&= b_0 = 1$$

$$&= b_1 = 4$$

$$&= b_2 = 6$$

$$&= b_3 = 4$$

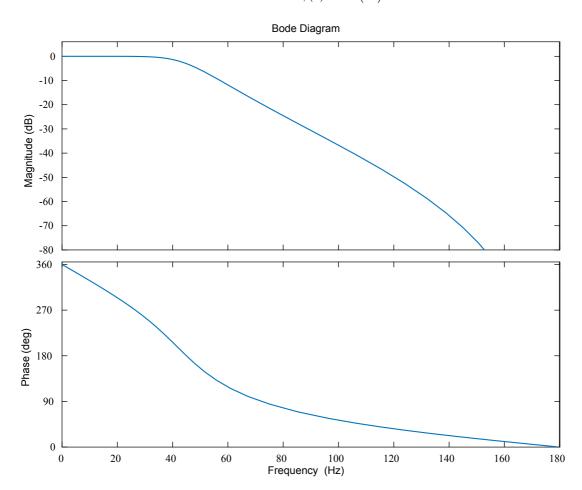
#### Frequency response

We can check the filter's frequency response to make sure that we didn't make any mistakes. As mentioned in other pages, the frequency response of a digital system can be a obtained by evaluating the transfer function H(z) along the unit circle  $(z=e^{j\omega})$ . We'll plot the magnitude in decibels

$$A(\omega) = 20 \log_{10} \lvert H\left(e^{j\omega}
ight) 
vert$$

We can also plot the phase angle of the response:

$$\phi(\omega) = \angle H\left(e^{j\omega}
ight)$$



You can see that the corner frequency lies around  $45~\mathrm{Hz}$ . We can check this mathematically:

$$A(\omega_{c,d}) = -3.01 \mathrm{dB}$$

#### **MATLAB**

If you have to design many different filters, you don't want to calculate them all by hand. Luckily, MATLAB and GNU Octave come with a command to calculate the coefficients of Butterworth filters.

```
\% Sample frequency in Hz \% Cut-off frequency in Hz
f_s = 360;
f^{-}c = 45;
                % Order of the butterworth filter
omega c = 2 * pi * f c;
                                  % Cut-off angular frequency
omega_c_d = omega_c / f_s;
                                  % Normalized cut-off frequency (digital)
[b, a] = butter(order, omega c d / pi);
                                                  % Design the Butterworth filter
disp("a = "); disp(a);
disp("b = "); disp(b);
                                                  % Print the coefficients
H = tf(b, a, 1 / f s);
                                                  % Create a transfer function
bode(H);
                                                  % Show the Bode plot
```

Note that MATLAB expects the normalized frequency as a number from 0 to 1, so we have to divide by  $\pi$  before passing it to the butter function

A similar function is available in the SciPy signal package: butter

```
from scipy.signal import butter, freqz
import matplotlib.pyplot as plt
from math import pi
import numpy as np
f s = 360
             # Sample frequency in Hz
             # Cut-off frequency in Hz
f c = 45
order = 4
             # Order of the butterworth filter
```

```
omega_c = 2 * pi * f_c # Cut-off angular frequency
omega_c_d = omega_c / f_s
                                   # Normalized cut-off pulsation (digital)
b, a = butter(order, omega_c_d / pi)
                                                # Design the Butterworth filter
print("a =", a)
print("b =", b)
                                                # Print the coefficients
w, h = freqz(b, a)
w *= f_s / (2 * pi)
plt.plot(w, 20 * np.log10(abs(h)))
plt.title('Frequency Response')
                                             # Calculate the frequency response
                                             # Convert from rad/sample to Hz
                                            # Convert to dB
                                             # Plot the result
plt.xlabel('Frequency [Hz]')
plt.ylabel('Magnitude [dB]')
plt.xlim(0, f_s / 2)
plt.ylim(-80, 6)
plt.axvline(f_c, color='red')
plt.show()
```

