# Digital Signal Processing IIR Filter Design via Impulse Invariance

D. Richard Brown III

#### Basic Procedure

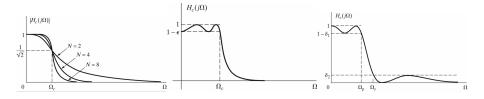
We assume here that we've already decided to use an IIR filter.

The basic procedure for IIR filter design via impulse invariance is:

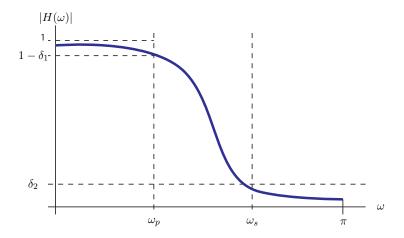
- 1. Determine the CT filter design method:
  - 1.1 Butterworth
  - 1.2 Chebychev Type I or Type II
  - 1.3 Elliptic
  - 1.4 ...
- 2. Transform the DT filter specifications to CT (sampling period  $T_d$  is arbitrary)
- 3. Design CT filter based on the magnitude squared response  $|H_c(j\Omega)|^2$ 
  - Determine filter order
  - Determine cutoff frequency
- 4. Determine  $H_c(s) \leftrightarrow h_c(t)$  corresponding to a stable causal filter
- 5. Convert to DT filter  $H(z)\leftrightarrow h[n]$  via impulse invariance such that  $h[n]=h_c(nT_d)$

# Determining the Continuous-Time Filter Design Method

	passband	stopband
Butterworth	monotonic	monotonic
Chebychev Type I	equiripple	monotonic
Chebychev Type II	monotonic	equiripple
equiripple	equiripple	equiripple



## Impulse-Invariant Lowpass Butterworth Filter Design Ex.



We start with the desired specifications of the DT filter. For this example, we will use  $\omega_p=0.2\pi$ ,  $\omega_s=0.3\pi$ ,  $1-\delta_1=0.89125$ , and  $\delta_2=0.17783$ .

### Convert DT Filter Specs to CT Filter Specs

We will use the Butterworth filter approach in this example. A CT Butterworth filter has a squared magnitude response given by

$$|H_c(j\Omega)|^2 = \frac{1}{1 + \left(\frac{j\Omega}{j\Omega_c}\right)^{2N}} \tag{1}$$

where  $\Omega_c$  is the cutoff frequency (radians/second) and N is the filter order.

When designing filters via impulse invariance, the sampling period  $T_d$  is arbitrary. It is often convenient to just set  $T_d=1$  so that  $\Omega=\omega$ .

This implies the CT filter specs can be written as  $\Omega_p=0.2\pi$  and  $\Omega_s=0.3\pi$ . Along with our magnitude specifications  $1-\delta_1=0.89125$  and  $\delta_2=0.17783$ , we can substitute these results directly into (1) to write

$$1 + \left(\frac{0.2\pi}{\Omega_c}\right)^{2N} \le \left(\frac{1}{0.89125}\right)^2$$
$$1 + \left(\frac{0.3\pi}{\Omega_c}\right)^{2N} \ge \left(\frac{1}{0.17783}\right)^2$$

#### Determine the N and $\Omega_c$

We can take the previous inequalities and write them as equalities as

$$1 + \left(\frac{0.2\pi}{\Omega_c}\right)^{2N} = \left(\frac{1}{0.89125}\right)^2 \tag{2}$$

$$1 + \left(\frac{0.3\pi}{\Omega_c}\right)^{2N} = \left(\frac{1}{0.17783}\right)^2. \tag{3}$$

We have two equations and two unknowns. By taking logarithms, we can isolate N and  $\Omega_c$  to get

$$N = 5.8858$$

$$\Omega_c = 0.70474$$

Note that N must be an integer, so we can choose N=6. We now can decide whether to pick  $\Omega_c$  to match the passband spec (and exceed the stopband spec) or match the stopband spec (and exceed the passband spec). To minimize the effect of aliasing, we usually choose the former.

# Determine $H_c(s)$ (part 1 of 3)

Given  ${\cal N}=6$  and our choice to match the passband spec, we have the equality

$$1 + \left(\frac{0.2\pi}{\Omega_c}\right)^{12} = \left(\frac{1}{0.89125}\right)^2 \tag{4}$$

which gives  $\Omega_c = 0.7032$ . Now, to determine  $H_c(s)$ , we can write

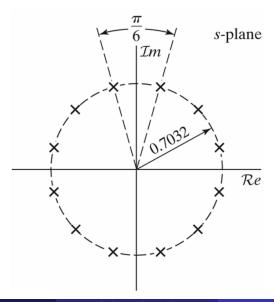
$$H_c(s)H_c(-s) = \frac{1}{1 + \left(\frac{s}{j\Omega_c}\right)^{2N}}.$$

The pole locations of  $H_c(s)H_c(-s)$  follow from the fact that  $1+\left(\frac{x}{a}\right)^M=0$  has M solutions given by

$$x = \{ae^{j\pi/M}, ae^{j3\pi/M}, \dots, ae^{j(\pi + (M-1)2\pi)/M}\}.$$

The  $\frac{M}{2}$  poles corresponding to  $H_c(s)$  are those in the left half plane.

# Determine $H_c(s)$ (part 2 of 3)



# Determine $H_c(s)$ (part 3 of 3)

Choosing the poles from the left half plane and doing a little bit of algebra, we can write

$$H_c(s) = \frac{0.12093}{(s^2 + 0.3640s + 0.4945)(s^2 + 0.9945s + 0.4945)(s^2 + 1.3585s + 0.4945)}$$

Since all of the poles are simple (non-repeated), we can write the partial fraction expansion

$$H_c(s) = \sum_{k=1}^{6} \frac{A_k}{s - s_k}$$

which implies that

$$h_c(t) = \begin{cases} \sum_{k=1}^6 A_k e^{s_k t} & t \ge 0\\ 0 & t < 0. \end{cases}$$

# Determine H(z) via Impulse Invariance

Impulse invariance requires  $h[n]=h_c(nT_d)$  where we have previously chosen  $T_d=1$ . Hence we have

$$h[n] = \begin{cases} \sum_{k=1}^{6} A_k e^{s_k n} & n \ge 0\\ 0 & t < 0. \end{cases}$$

and it follows that

$$H(z) = \sum_{k=1}^{6} \frac{A_k}{1 - e^{s_k} z^{-1}}$$

with ROC  $\left|z\right|>$  largest magnitude pole. A bit of algebra yields the final result

$$H(z) = \frac{0.2871 - 0.4466z^{-1}}{1 - 1.2971z^{-1} + 0.6949z^{-2}} + \frac{-2.1428 + 1.1455z^{-1}}{1 - 1.0691z^{-1} + 0.3699z^{-2}} + \frac{1.8577 - 0.6303z^{-1}}{1 - 0.9972z^{-1} + 0.2570z^{-2}}$$

which can immediately be realized in parallel form or rearranged to be realized in cascade or direct forms.

### Impulse-Invariant Lowpass Butterworth Filter Design Ex.

