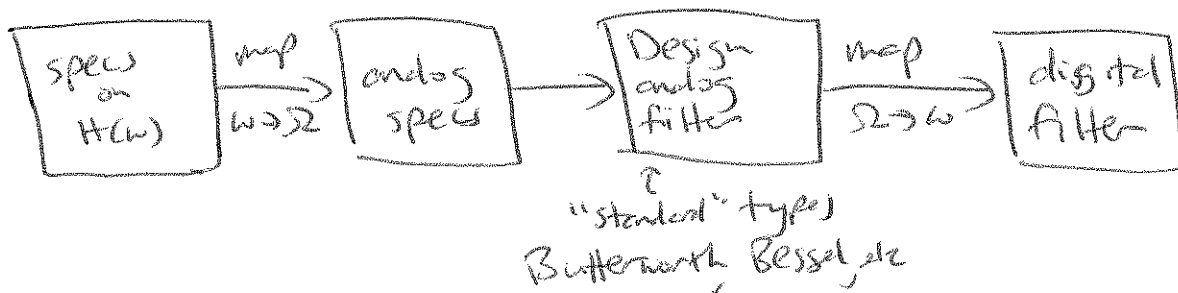


IIR Filter Design

(POM 10.3.3)

Basic idea

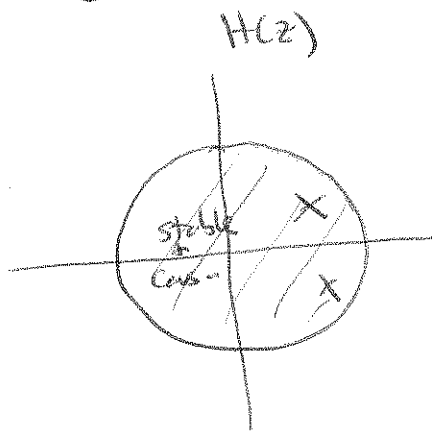
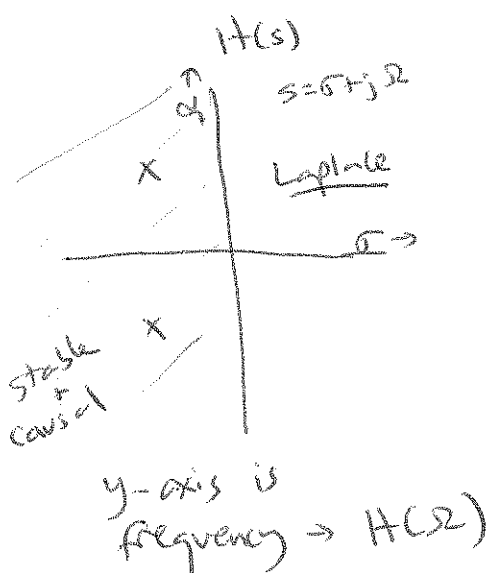
- 1) state specifications for $H(\omega)$
- 2) map specs to continuous frequency ($\omega \Rightarrow \Omega$)
 $\Omega = 2\pi f$
- 3) design a CT (analog) filter that meets spec
- 4) transform back to DT filter ($\Omega \Rightarrow \omega$)



This seems strange. Why do it?

- there are many really good, well-understood analog filters
- because filters are well understood, no need for iterative design

Problem: how to map CT freq response $H(s)$ (Laplace) to DT " $H(z)$ "?



unit circle is frequency $\rightarrow H(\omega)$

Three main mappings

- 1) approx by derivatives
- 2) impulse invariance
- 3) bilinear transform (best)

note: Ω is ∞ long
 ω is 2π long

Approximation by derivative (1)

Idea: 1) represent a derivative in continuous time vs discrete time

2) transform each into freq domain

3) get a mapping

$$\frac{dy}{dt} \approx \frac{y(nT) - y(nT-T)}{T} = \frac{\Delta y}{\Delta t} \quad T = \text{sampling interval}$$

$$\frac{dy}{dt} \approx \frac{y(n) - y(n-1)}{T}$$

z transform

Laplace transform

$$sY(z) \approx \frac{1}{T} (Y(z) - z^{-1}Y(z)) = \frac{1-z^{-1}}{T} Y(z)$$

or, map

$$s = \frac{1-z^{-1}}{T}$$

map s to z

So approximate: $H(z) = H(s) \big|_{s = \frac{1-z^{-1}}{T}}$ plus in for s

where do CT freq map to?

$$s = \frac{1-z^{-1}}{T}$$

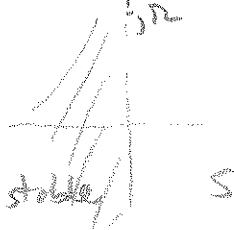
\Rightarrow

$$z = \frac{1}{1-Ts}$$

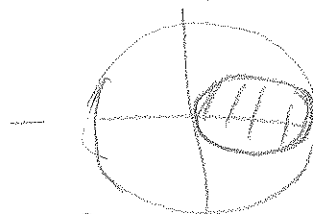
\rightarrow so on jR axis,

$$z = \frac{1}{1-Tj\Omega}$$

$$= \frac{1}{1+R^2T^2} + j \frac{RT}{1+R^2T^2}$$



stable



stable but limited

$$\text{at } R=0, z = \frac{1}{1} = 1$$

$$\text{as } R \rightarrow \pm\infty, z \rightarrow \frac{1}{\pm\infty} = 0$$

in between, maps over circle

Approx by derivative - 2

ex) say we have analog filter

$$H_a(s) = \frac{1}{(s+0.1)^2 + 9}$$

this mapping gives

$$H(z) = \frac{1}{\left(\frac{1-z^{-1}}{T} - 0.1\right)^2 + 9}$$

We start with a system (in Laplace-land)

$$H(s) = \frac{b}{s + a}$$

but $H(s) = Y(s)/X(s)$, so

$$Y(s)(s + a) = bX(s)$$

As you hopefully remember from a linear systems class, 's' in the Laplace transform corresponds to a time derivative, so in the time domain we get

$$y'(t) + ay(t) = bx(t) \quad (1)$$

In general, it's true from calculus that

$$y(t) = y(t_0) + \int_{t_0}^t y'(\tau) d\tau$$

We approximate the integral by the trapezoidal rule, and evaluate at the points $t = nT$ and $t_0 = (n-1)T$ (notice that by moving to time sampled at intervals of T , we are moving from continuous to sampled time). Then, the above becomes

$$y(nT) \approx y(nT - T) + T \left(\frac{y'(nT) + y'(nT - T)}{2} \right) \quad (2)$$

However, Eq. 1 gives a result for the derivative. We can plug this into Eq 2, and do a bunch of algebra. If we define $y(n) = y(nT)$ for simplicity, we have

$$(1 + aT/2)y(n) - (1 - aT/2)y(n-1) = bT/2(x(n) + x(n-1))$$

If we take the z-transform (as we are now in discrete time), we get

$$(1 + aT/2)Y(z) - (1 - aT/2)z^{-1}Y(z) = bT/2(1 + z^{-1})X(z)$$

Collecting $Y(z)$ terms and rearranging, we can find

$$H(z) = \frac{Y(z)}{X(z)} = \frac{b}{\frac{2}{T} \frac{1-z^{-1}}{1+z^{-1}} + a}$$

but

$$H(z) \approx H(s) = \frac{b}{s + a}$$

(the reason it's approximate is that the trapezoidal rule only approximated the integral). Thus

$$s \approx \frac{2}{T} \frac{1 - z^{-1}}{1 + z^{-1}}$$

which is the bilinear transform.

(2)
Bilinear transform \rightarrow see pdf for derivation

get
$$s = \frac{z}{T} \frac{1 - z^{-1}}{1 + z^{-1}}$$

where $T = 1/F_s$

Characteristics

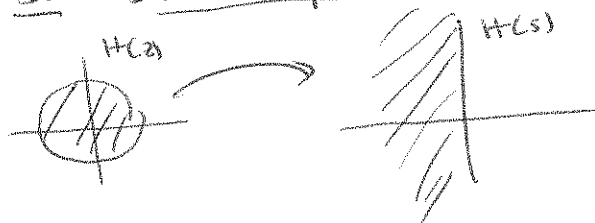
First, we write $z = r e^{j\omega}$
 $s = \sigma + j\Omega$

after algebra, (see book) get

$$\left[\begin{array}{l} a) \quad \sigma = \frac{z}{T} \frac{r^2 - 1}{1 + r^2 + 2r \cos \omega} \\ b) \quad \Omega = \frac{z}{T} \frac{2r \sin \omega}{1 + r^2 + 2r \cos \omega} \end{array} \right.$$

from a) If $r < 1$, then $\sigma < 0$; if $r > 1$, $\sigma > 0$

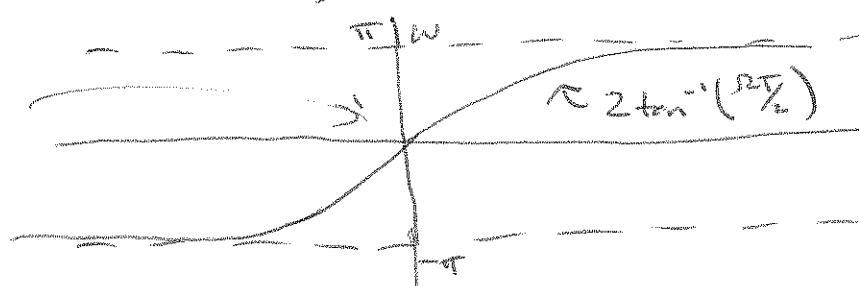
So stable maps to stable



from b), look at freq response, on unit circle, $r=1$

$$\Omega = \frac{z}{T} \frac{2 \sin \omega}{2 + 2 \cos \omega} = \frac{z}{T} \frac{\sin \omega}{1 + \cos \omega} = \frac{z}{T} \tan \frac{\omega}{2}$$

$$\text{or } \omega = 2 \tan^{-1}(\Omega T/2)$$



as $T \downarrow$ ($F_s \uparrow$)
 a given Ω
 maps closer
 to linear
 region

nearly
 linear
 near
 $\Omega T = 0$

(3)

Some examples

Case 1 : specify { general form for $H(s)$
specifications in ω

T will divide out

ex / $\rightarrow H(s) = \frac{\Omega_c}{s + \Omega_c}$ $\Omega_c = -3\text{dB point}$
 form lowpass
 problem statement \rightarrow say we want digital cutoff (-3 dB point)
 at 0.2π ; $\omega_c = 0.2\pi$

step 1) warp ω_c to Ω_c

$$\Omega_c = \frac{2}{T} \tan \frac{\omega_c}{2} = \frac{2}{T} \tan \frac{0.2\pi}{2} = \frac{0.65}{T}$$

$$\text{so } H(s) = \frac{\Omega_c}{s + \Omega_c} = \frac{0.65/T}{s + 0.65/T}$$

$$\text{bilinear map: } s = \frac{z}{T} \frac{1-z^{-1}}{1+z^{-1}}$$

$$H(z) = \frac{0.65/T}{\frac{z}{T} \left(\frac{1-z^{-1}}{1+z^{-1}} \right) + 0.65/T} = \frac{0.65}{2 \left(\frac{1-z^{-1}}{1+z^{-1}} \right) + 0.65}$$

T divides out

now we could find poles/zeros etc.

(4)

Case 2 Sometimes $H(s)$ is specified - not just in general terms (Ω_c, etc) but in more detail (say, CT frequencies are fixed). \nwarrow trying to translate a specific filter, not a type of filter. Then, we need to pick T to match our digital spec.

a) $\left\{ \begin{array}{l} \text{given } H(s) = \frac{s+0.1}{(s+0.1)^2 + 16} \\ \text{problem statement} \end{array} \right.$ and want resonant freq at $\omega = \pi/2$

this analog filter is a resonator w/ resonance at

$$\Omega = 4, \text{ or}$$

$$2\pi F = 4 \\ F = \frac{4}{2\pi} = \frac{2}{\pi} \text{ Hz}$$

the mapping is still

$$\Omega = \frac{2}{T} \tan \frac{\omega}{2}$$

but now Ω & ω are set; find T

$$4 = \frac{2}{T} \tan\left(\frac{\pi/2}{2}\right) \Rightarrow T = \frac{2}{4} \tan\left(\frac{\pi}{4}\right) \\ T = \frac{1}{2} \quad \underline{F = 2 \text{ Hz}}$$

then,
$$s = \frac{2}{T} \left(\frac{1-z^{-1}}{1+z^{-1}} \right) = 4 \left(\frac{1-z^{-1}}{1+z^{-1}} \right)$$

plug in:

$$H(z) = \frac{4 \left(\frac{1-z^{-1}}{1+z^{-1}} \right) + 0.1}{\left(4 \left(\frac{1-z^{-1}}{1+z^{-1}} \right) + 0.1 \right)^2 + 16}$$

Simplify, find poles/zeros
inverse z transform to get $h[n]$

(5)

In matlab (or similar), this is easy:

$[b, a] = \text{butter}(n, W_n)$ for low pass

$n = \text{filter order}$

$W_n = \text{normalized freq: } W_n = f_c / (f_s / 2)$
 \uparrow
 cutoff

Q) how to pick n ?

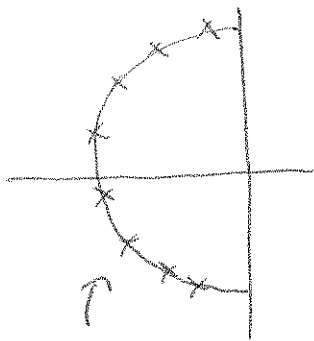
A) use 'butterord' command

Q) what does that do?

A) look at Butterworth response:

$$|H(\Omega)|^2 = \frac{1}{1 + (\Omega/\Omega_c)^{2n}} = \frac{1}{1 + \epsilon^2 (\Omega/\Omega_p)^{2n}}$$

\uparrow
specific form of n^{th} order Butterworth \uparrow



Butterworth poles are on unit circle in s

$\Omega_c = -3 \text{ dB cutoff frequency}$

$\Omega_p = \text{passband frequency}$

alternate spec $\left(\frac{1}{1 + \epsilon^2} \right)$ is magnitude when $\Omega = \Omega_p$

to meet an attenuation δ_2 at some frequency Ω_s
 \uparrow
 stop freq

$$\frac{1}{1 + \epsilon^2 (\Omega_s/\Omega_p)^{2n}} = \delta_2^2$$

\uparrow if we specify $\delta_2, \epsilon, \Omega_s, \Omega_p$, we can solve for n that is big enough to meet spec.

→ Similar formulas for other filters

- 4) You are trying to build a digital filter to mimic a simple RC lowpass filter. This RC filter has the Laplace transform response:

$$H(s) = (1/RC) / (s + (1/RC))$$

i.e., it is lowpass with cutoff frequency $\Omega_c = 1/RC$. The filter you are trying to match has a 1000 ohm resistor and a 1e-6 Farad capacitor.

- a) What is the cutoff frequency, in Hz, of the RC filter? If your digital system has a sampling frequency of 3000 Hz, what radian frequency ω_c does this correspond to?

$$F_c = \frac{1}{2\pi RC} = \frac{1000}{2\pi} \text{ Hz}$$

$$\omega_c = 2\pi \frac{F_c}{F_s} = 2\pi \frac{1000}{2\pi \cdot 3000} = \frac{2\pi}{3} \cdot \frac{1}{2\pi} = \frac{1}{3}$$

- c) Assuming you are using the bilinear transform to do the mapping, find the Ω_c corresponding to ω_c .

$$\Omega_c = \frac{2}{T} \tan \frac{\omega_c}{2} = \frac{2}{T} \tan \frac{1/3}{2} \approx 1009$$

- d) take the first step in finding $H(z)$ – you do not need to do any algebraic simplifications.

plug in bilinear transform

$$s = \frac{2}{T} \left(\frac{1-z^{-1}}{1+z^{-1}} \right)$$

$$H(z) = \frac{\Omega_c}{\frac{2}{T} \left(\frac{1-z^{-1}}{1+z^{-1}} \right) + \Omega_c}$$