Digital Signal Processing MATLAB HW3 - q2

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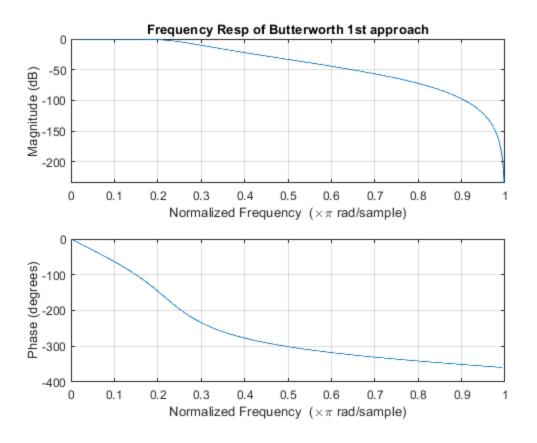
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Clear recent data

```
clear; close all; clc;
```

A. Bilinear transformation

```
part 1
% define parameters
wp = 0.2*pi;
ws = 0.4*pi;
Rp = 1;
As = 20;
T = 2;
Fs = 1/T
OmegaP = (2/T)*tan(wp/2);
OmegaS = (2/T)*tan(ws/2);
[cs,ds] = afd_butt(OmegaP,OmegaS,Rp,As); %butterworth filter design in
 continuous domain
[b1,a1] = bilinear(cs,ds,Fs); %butterworth continuous to discrete
figure(1);
freqz(b1,a1);
title('Frequency Resp of Butterworth 1st approach');
Fs =
    0.5000
*** Butterworth Filter Order = 4
```

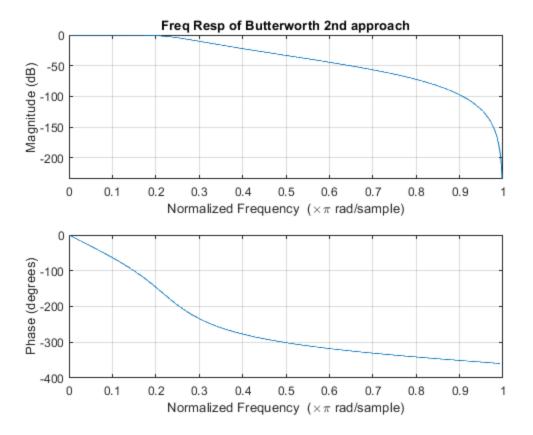


B. Normalized Butterworth

```
part 2

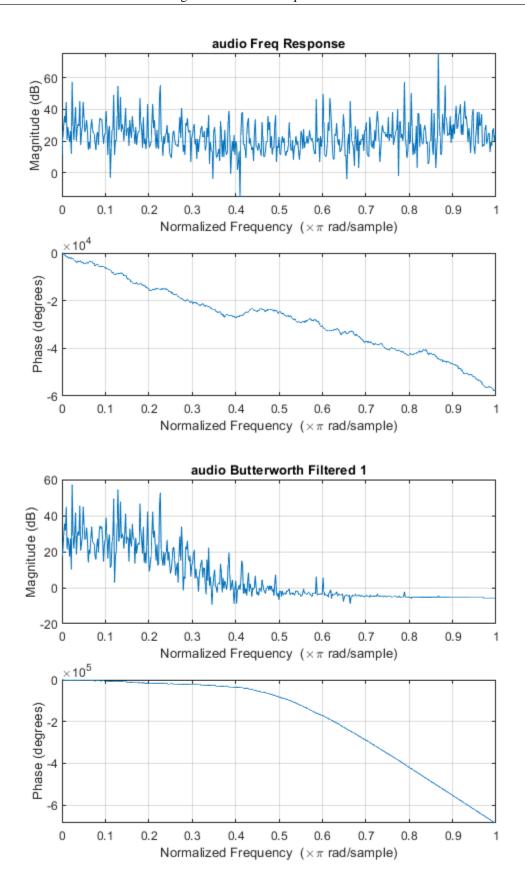
OmegaP = (2/T)*tan(wp/2);
OmegaS = (2/T)*tan(ws/2);
N =ceil((log10((10^(Rp/10)-1)/(10^(As/10)-1))))/(2*log10(OmegaP/OmegaS)));
OmegaC = OmegaP/((10^(Rp/10)-1)^(1/(2*N)));

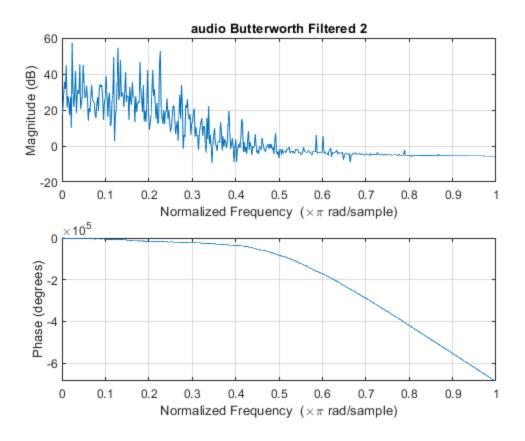
wn = (2/pi)*atan((OmegaC*T)/2);
[b2,a2] = butter(N,wn); %butterworth direct design in discrete domain figure(2);
freqz(b2,a2);
title('Freq Resp of Butterworth 2nd approach');
```



C.Filtering

```
part 3
[audio , Fs] = audioread('HW3_Q2_multi_tone.wav');
audio = (audio)';
                                        %transposing audio file
figure(3);
freqz(audio);
title('audio Freq Response');
filtered_audio1 = filter(b1,a1,audio); %first methof filter on audio
file
figure(4);
freqz(filtered_audio1);
title('audio Butterworth Filtered 1');
filtered_audio2 = filter(b2,a2,audio); %second method filter on audio
file
figure(5);
freqz(filtered_audio2);
title('audio Butterworth Filtered 2');
```





Butterworth Function

```
function [b,a] = u_buttap(N,Omegac) %%copy butterworth design function
 from the book
[z,p,k] = buttap(N);
p = p*Omegac;
k = k*Omegac^N;
B = real(poly(z));
b0 = k;
b = k*B;
a = real(poly(p));
end
function [b,a] = afd_butt(Wp,Ws,Rp,As)
if Wp <= 0
error('Passband edge must be larger than 0')
end
if Ws <= Wp
error('Stopband edge must be larger than Passband edge')
end
if (Rp <= 0) | (As < 0)
error('PB ripple and/or SB attenuation ust be larger than 0')
end
N = ceil((log10((10^{(Rp/10)-1)}/(10^{(As/10)-1)}))/(2*log10(Wp/Ws)));
fprintf('\n*** Butterworth Filter Order = %2.0f \n',N);
OmegaC = Wp/((10^{(Rp/10)-1)^{(1/(2*N))}};
```

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[b,a]=u_buttap(N,OmegaC);
end

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