Digital Signal Processing MATLAB HW2 - q2

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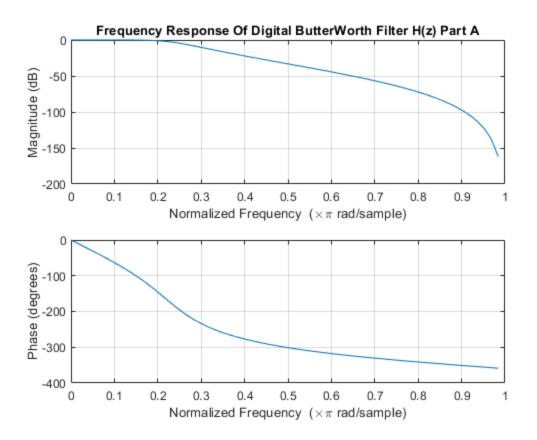
Professor: Dr. Sheikhzadeh Author: [SeyedAli] - [SeyedHosseini] E-mail: [alisnake@aut.ac.ir] % which Γ m about to change ASAP University: Amirkabir University of Technology

Clear recent data

clear; close all; clc;

Part A

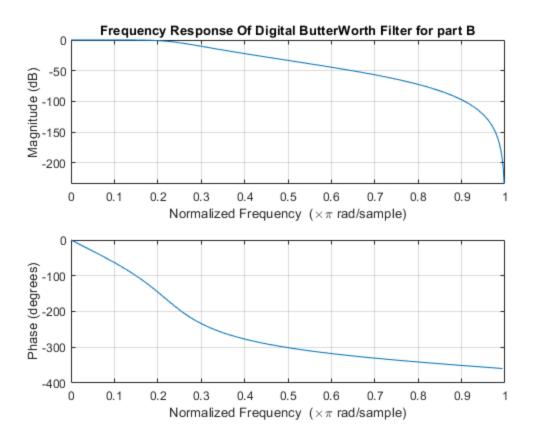
```
% Digital Filter Specifications:
wp = 0.2*pi; % digital Passband freq in Hz
ws = 0.4*pi; % digital Stopband freq in Hz
Rp = 1; % Passband ripple in dB
As = 20; % Stopband attenuation in dB
% Analog Prototype Specifications:
T = 2; % Set T=2
OmegaP = (2/T)*tan(wp/2); % Prewarp Prototype Passband freq
OmegaS = (2/T)*tan(ws/2); % Prewarp Prototype Stopband freq
[cs,ds] = afd_butt(OmegaP,OmegaS,Rp,As); % Analog Butterworth
Prototype Filter Calculation
Fs = 1/T; %sample rate to a discrete equivalent.
[cz , dz] = bilinear(cs,ds,Fs); %Bilinear Transformation on H(s) to
H(z)
[C,B,A] = sdir2cas(cz,dz); %FInding Coefficient In Cascade Form
figure(1)
freqz(cz,dz,64) %plotting Freq Response of H(z)
title("Frequency Response Of Digital ButterWorth Filter H(z) Part A")
```



Part B

```
% Digital Filter Specifications:
wp = 0.2*pi; % digital Passband freg in Hz
ws = 0.4*pi; % digital Stopband freq in Hz
Rp = 1; % Passband ripple in dB
As = 20; % Stopband attenuation in dB
% Analog Prototype Specifications:
T = 2; % Set T=2
OmegaP = (2/T)*tan(wp/2); % Prewarp Prototype Passband freq
OmegaS = (2/T)*tan(ws/2); % Prewarp Prototype Stopband freq
% Analog Prototype Order Calculation:
N = ceil((log10((10^(Rp/10)-1)/(10^(As/10)-1)))/(2*log10(OmegaP/
OmegaS)));
fprintf('\n Butterworth Filter Order for Part B = %2.0f \n',N)
OmegaC_b = OmegaP/((10^{(Rp/10)-1)^(1/(2*N)));  %Analog BW prototype
 cutoff
wn = 2*atan((OmegaC_b*T)/2); %Digital BW cutoff freq
% Digital Butterworth Filter Design:
wn = wn/pi; %Digital Butter cutoff in pi units
[czb,dzb]=butter(N,wn);
[Cb,Bb,Ab] = sdir2cas(czb,dzb); %FInding Coefficient In Cascade Form
```

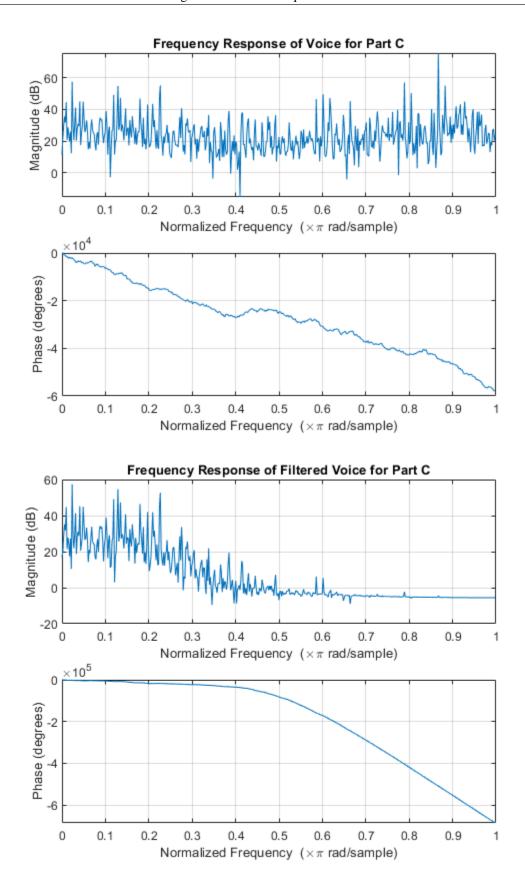
```
figure(2)
freqz(czb,dzb)
title("Frequency Response Of Digital ButterWorth Filter for part B")
Butterworth Filter Order for Part B = 4
```



Part C

```
[voice , Fs1] = audioread("multi_tone.wav"); %Storing audio in matlab
voice = voice';
figure(3)
freqz(voice)
title(" Frequency Response of Voice for Part C ")

voice_filtered = filter(czb,dzb,voice); %Filtered Voice with Designed
Filtered in Section B
figure(4)
freqz(voice_filtered)
title(" Frequency Response of Filtered Voice for Part C ")
```



Function Impulse Invariance

```
function [b,a] = imp_invr(c,d,T)
% Impulse Invariance Transformation from Analog to Digital Filter
% ------
% [b,a] = imp_invr(c,d,T)
% b = Numerator polynomial in z^(-1) of the digital filter
% a = Denominator polynomial in z^(-1) of the digital filter
% c = Numerator polynomial in s of the analog filter
% d = Denominator polynomial in s of the analog filter
% T = Sampling (transformation) parameter
%
[R,p,k] = residue(c,d); p = exp(p*T);
[b,a] = residuez(R,p,k); b = real(b'); a = real(a');
end
```

Function for ButterWorth Design1

Function for ButterWorth Design2

```
function [b,a] = afd_butt(Wp,Ws,Rp,As);
% Analog Lowpass Filter Design: Butterworth
% -------
% [b,a] = afd_butt(Wp,Ws,Rp,As);
% b = Numerator coefficients of Ha(s)
% a = Denominator coefficients of Ha(s)
% Wp = Passband edge frequency in rad/sec; Wp > 0
% Ws = Stopband edge frequency in rad/sec; Ws > Wp > 0
% Rp = Passband ripple in +dB; (Rp > 0)
% As = Stopband attenuation in +dB; (As > 0)
%
if Wp <= 0
error('Passband edge must be larger than 0')
end
if Ws <= Wp</pre>
```

```
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 = \text{ceil}((\log 10((10^{(Rp/10)-1)/(10^{(As/10)-1)}))/(2*\log 10(Wp/Ws)))}; \\ \text{fprintf('} N \text{ Butterworth Filter Order for Part A = $2.0f } n', N) \\ \text{OmegaC} = Wp/((10^{(Rp/10)-1)^{(1/(2*N))});} \\ \text{[b,a]=u_buttap(N,OmegaC)}; \\ \text{end}
```

Butterworth Filter Order for Part A = 4

Function Direct to Cascade Form

```
function [C,B,A] = sdir2cas(b,a);
% DIRECT-form to CASCADE-form conversion in s-plane
% ______
% [C,B,A] = sdir2cas(b,a)
% C = gain coefficient
% B = K by 3 matrix of real coefficients containing bk's
% A = K by 3 matrix of real coefficients containing ak's
% b = numerator polynomial coefficients of DIRECT form
% a = denominator polynomial coefficients of DIRECT form
Na = length(a)-1; Nb = length(b)-1;
% compute gain coefficient C
b0 = b(1); b = b/b0; a0 = a(1); a = a/a0; C = b0/a0;
% Denominator second-order sections:
p= cplxpair(roots(a)); K = floor(Na/2);
if K*2 == Na % Computation when Na is even
  A = zeros(K,3);
   for n=1:2:Na
      Arow = p(n:1:n+1,:); Arow = poly(Arow);
      A(fix((n+1)/2),:) = real(Arow);
   end
 elseif Na == 1 % Computation when Na = 1
  A = [0 \text{ real(poly(p))}];
 else % Computation when Na is odd and > 1
   A = zeros(K+1,3);
    for n=1:2:2*K
      Arow = p(n:1:n+1,:); Arow = poly(Arow);
      A(fix((n+1)/2),:) = real(Arow);
   A(K+1,:) = [0 real(poly(p(Na)))];
  end
% Numerator second-order sections:
   z = cplxpair(roots(b)); K = floor(Nb/2);
   if Nb == 0 % Computation when Nb = 0
      B = [0 \ 0 \ poly(z)];
   elseif K*2 == Nb % Computation when Nb is even
```

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```
B = zeros(K,3);
      for n=1:2:Nb
          Brow = z(n:1:n+1,:); Brow = poly(Brow);
          B(fix((n+1)/2),:) = real(Brow);
      end
   elseif Nb == 1 % Computation when Nb = 1
        B = [0 real(poly(z))];
     else % Computation when Nb is odd and > 1
       B = zeros(K+1,3);
        for n=1:2:2*K
Brow = z(n:1:n+1,:); Brow = poly(Brow);
B(fix((n+1)/2),:) = real(Brow);
        end
           B(K+1,:) = [0 real(poly(z(Nb)))];
   end
end
```

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