# **FINAL PROJECT - ECES 631**

#### **Table of Contents**

DISCRETE TIME MODELS FOR THE SPEECH SIGNAL	1
Part 3.3 a	1
Part 3.3 b	1
Part 3.3c	1

# DISCRETE TIME MODELS FOR THE SPEECH SIGNAL

AUTHOR: SUNDAR RAM

## **Part 3.3 a**

#### Exponential model

```
a=0.91;
Npts=51;
Nfreq=6;
[gE,GE,w_E] = glottalE(a,Npts,Nfreq);
% figure,plot(gE), title('Exponential Model');
% Rosenberg Model
N1=40;
N2=10;
[gR,GR,w_R] = glottalR(N1,N2,Nfreq);
```

## Part 3.3 b

#### FLIPPED ROSENBERG

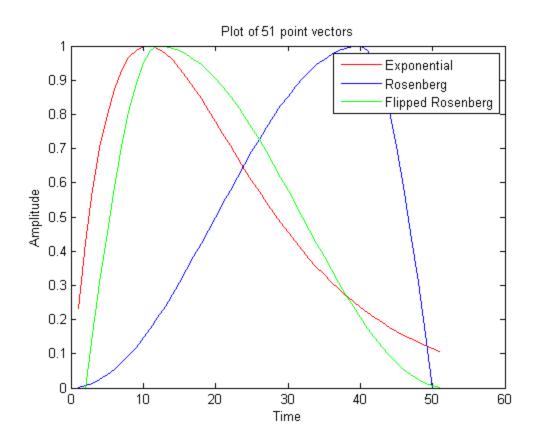
```
gRflip = fliplr(gR);
GR_flip=zeros(1,length(w_R));

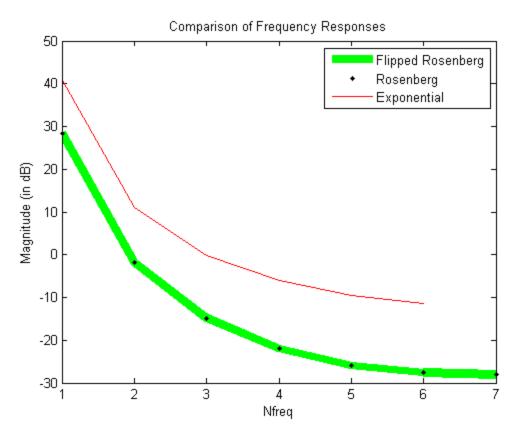
for p=1:length(w_R)
    win=w_R(p);
    for l=1:N1+N2
        cmn=0;
%        cmn=gRflip(l)*exp((1i)*p*2*pi/Nfreq)*1;
        cmn=gR(l)*exp(-(1i)*win*1);
        GR_flip(p)=GR_flip(p) + cmn;
    end
end
```

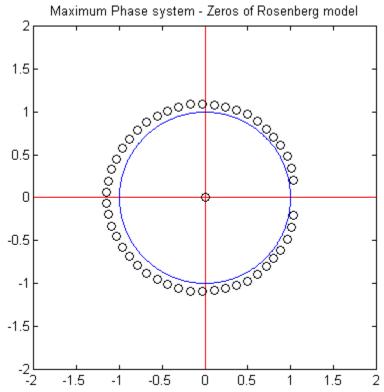
# Part 3.3c

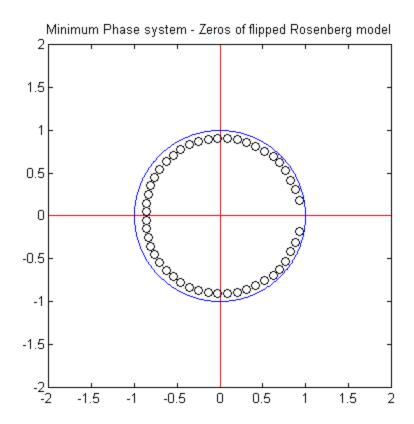
```
gE=gE/max(gE);
```

```
figure, plot(gE,'r');
hold on, plot(qR, 'b');
hold on, plot(gRflip,'g');
title('Plot of 51 point vectors')
legend('Exponential','Rosenberg','Flipped Rosenberg');
xlabel('Time'), ylabel('Amplitude');
%Frequency plots
figure,plot(20*log10(abs(GR_flip)),'g','linewidth',6);
hold on, plot(20*log10(abs(GR)),'k.'), title('Comparison of Frequency Responses');
plot(20*log10(abs(GE)), 'r');
legend('Flipped Rosenberg','Rosenberg','Exponential');
xlabel('Nfreq'), ylabel('Magnitude (in dB)');
%%Part 3.3d
% For the rosenberg model, since there is a zero at z = 0, when we flip it,
% then the zero goes to infinity. So skipping the first zero and plotting
% the rest of the zeros for the flipped rosenberg model
roo1=roots(qR);
figure,zpl(roo1,[]),title('Maximum Phase system - Zeros of Rosenberg model');
roo2=roots(qRflip);
figure, zpl(roo2(2:end),[]), title('Minimum Phase system - Zeros of flipped Rosenber
```









```
function [gE,GE,w_E] = glottalE(a,Npts,Nfreq)
% gE is the exponential glottal waveform vector of length Npts
% GE is the frequency response at Nfreq frequencies between 0 and pi
% radians
gE=[];
GE=[];

x=1:Npts;
gE=[gE x.*a.^x];
% G(z) = az^-1 /(1-2az^-1+a^2z^-2) => b=a, a=[1 -2a a^2]

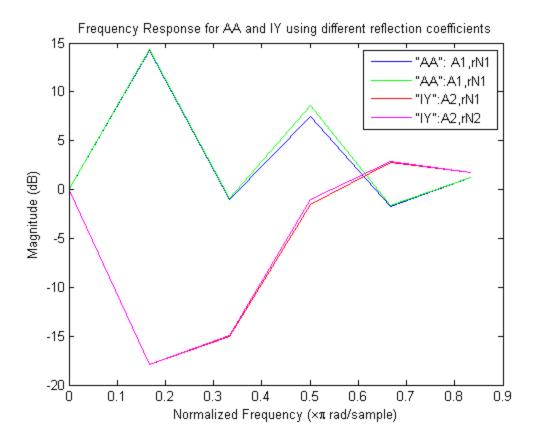
b = [0 a];
a0 = [1 -2*a a^2];
[GE,w_E]=freqz(b,a0,Nfreq);
end
```

```
function [gR,GR,w_R] = glottalR(N1,N2,Nfreq)
% gR is the Rosenberg Glottal waveform vector of length N1+N2+1
% GR is the frequency response at Nfreq frequencies w_R between 0 and pi
% radians
gR=[];
%Generation of glottal waveform
for n=1:N1+N2+1
    if(n<=N1)
    gR(n)=0.5*(1-cos(pi*(n)/N1));
    elseif(n<=N1+N2)</pre>
    gR(n) = cos(pi*(n-N1)/(2*N2));
    else
    gR(n) = 0;
    end
end
Frequency response of the glottal waveform using Fourier transform formula
w_R = 0:pi/Nfreq:pi;
GR=zeros(1,length(w_R));
for p=1:length(w_R)
    win=w_R(p);
    for l=1:N1+N2
        cmn=0;
          cmn=gR(1)*exp(-(1i)*p*2*pi/Nfreq)*1;
         cmn=gR(1)*exp(-(1i)*win*1);
        GR(p)=GR(p) + cmn;
    end
end
end
```

## **Part 4.1**

#### Frequency response plots

```
Nfreq = 6;
A1=[1.6,2.6,0.65,1.6,2.6,4.0,6.5,8.0,7.0,5.0];
A2=[2.6,8.0,10.5,10.5,8.0,4.0,0.65,0.65,1.3,3.2];
rN1=0.71;
rN2=1.0;
[r1,D1,G1]=atov(A1,rN1);
[h1,w1]=freqz(G1,D1,Nfreq);
figure,plot(w1/pi,20*log10(abs(h1)));
hold on;
[r2,D2,G2]=atov(A1,rN2);
[h2,w2]=freqz(G2,D2,Nfreq);
plot(w2/pi,20*log10(abs(h2)),'g');
[r3,D3,G3]=atov(A2,rN1);
[h3,w3]=freqz(G3,D3,Nfreq);
plot(w3/pi,20*log10(abs(h3)),'r');
[r4,D4,G4] = atov(A2,rN2);
[h4,w4] = freqz(G4,D4,Nfreq);
plot(w4/pi,20*log10(abs(h4)),'m');
legend('"AA": A1,rN1','"AA":A1,rN1','"IY":A2,rN1','"IY":A2,rN2');
xlabel('Normalized Frequency (\times\pi rad/sample)');
ylabel('Magnitude (dB)');
title('Frequency Response for AA and IY using different reflection coefficients');
```

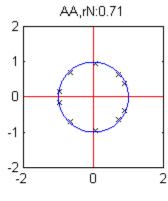


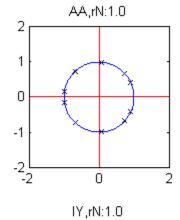
# Part 4.1b

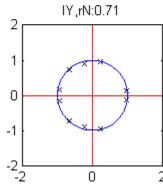
#### Pole plot

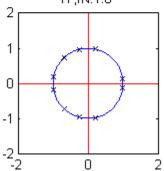
```
figure;
roo1 = roots(D1);
subplot(2,2,1), zpl([],roo1), title('AA,rN:0.71');
roo2 = roots(D2);
subplot(2,2,2), zpl([],roo2), title('AA,rN:1.0');
roo3 = roots(D3);
subplot(2,2,3), zpl([],roo3), title('IY,rN:0.71');
roo4 = roots(D4);
subplot(2,2,4), zpl([],roo4), title('IY,rN:1.0');
% FORMAT Frequencies
%Given T=1/1000;
Ts=1/1000;
ang1 = angle(roo1);
ang2 = angle(roo2);
ang3 = angle(roo3);
ang4 = angle(roo4);
```

```
fre1 = ang1/(2*pi*Ts);
fre2 = ang2/(2*pi*Ts);
fre3 = ang3/(2*pi*Ts);
fre4 = ang4/(2*pi*Ts);
```









Published with MATLAB® R2013a

### **Table of Contents**

System Models	. 1
Rosenberg Flipped	. 2
Part 5.1	
Part 5.2	
Part 5.3	

# **System Models**

```
%Exponential glottal pulse
fs=10000;
b = [0 \ a];
a0 = [1 - 2*a a^2];
sys_G_z = tf(b,a0,1/fs,'variable','z^-1');
% Vocal tract V(z)
sys_V_z1 = tf(G1,D1,1/fs,'variable','z^-1');
sys_V_z^2 = tf(G_2,D_2,1/fs,'variable','z^-1');
sys V z3 = tf(G3,D3,1/fs,'variable','z^-1');
sys_V_z4 = tf(G4,D4,1/fs,'variable','z^-1');
R(z) = (1-z^-1)
a=1;
b=[1 -1];
sys_R_z = tf(b,a,1/fs,'variable','z^-1');
% H(z) = G(z)*V(z)*R(z) for Exponential glottal model
H_z1 = sys_G_z*sys_V_z1*sys_R_z;
H_z2 = sys_G_z*sys_V_z2*sys_R_z;
H z3 = sys G z*sys V z3*sys R z;
H_z4 = sys_G_z*sys_V_z4*sys_R_z;
%%Excitation signal e[n] = y1
f = 100;
fs=10000;
y1=zeros(10000,1);
y1(1:fs/f:end)=1;
% stem(y1);
%Inverse Z-transform
imp1 = impz(H z1.num{1}, H z1.den{1}, 10);
imp2 = impz(H_z2.num{1}, H_z2.den{1},10);
imp3 = impz(H_z3.num{1}, H_z3.den{1}, 10);
imp4 = impz(H_z4.num{1}, H_z4.den{1}, 10);
%Convolution of input excitation signal with H(z)
soun_exp1=conv(y1,imp1);
soun_exp2=conv(y1,imp2);
soun_exp3=conv(y1,imp3);
soun_exp4=conv(y1,imp4);
```

```
% figure, subplot(2,2,1), plot(soun exp1),title('AA,rN:0.71');
 %Concatenate sounds into a single pulse.
song = cat(1, soun_exp1, zeros(1000, 1), soun_exp2, zeros(1000, 1), soun_exp3, zeros(1000, 1), soun_
 %Listen to the sounds
 soundsc(song,fs);
 % Rosenberg Glottal Pulse
 %FOR "AA" sound
 convolVR=sys_R_z*sys_V_z1;
 %Inverse of convolVR_z
himp = impz(convolVR.num{1},convolVR.den{1},10);
ans w=conv(himp,qR);
sou_n=conv(y1,ans_w);
 soundsc(sou n,fs);
 %FOR 'IY' sound
convolVR3=sys_R_z*sys_V_z3;
 %Inverse of convolVR z
himp3 = impz(convolVR3.num{1},convolVR3.den{1},10);
ans_w3=conv(himp3,gR);
sou_n3=conv(y1,ans_w3);
 soundsc(sou n3,fs);
r_song = cat(1, sou_n, zeros(1000, 1), sou_n3);
 soundsc(r_song);
```

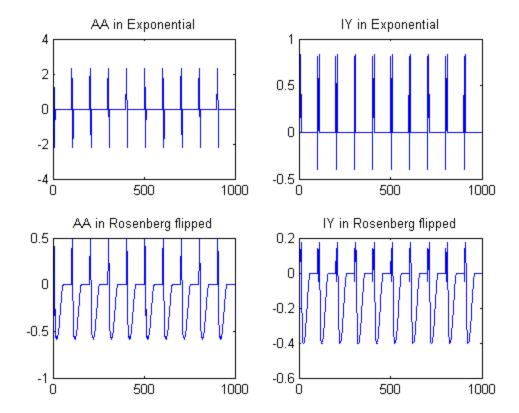
# Rosenberg Flipped

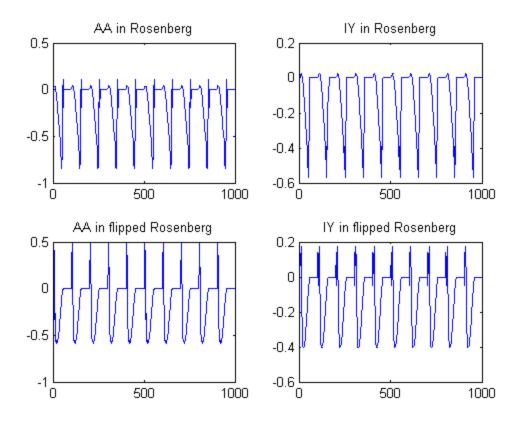
```
%FOR "AA" sound
ans_w_f=conv(himp,gRflip);
sou_nf=conv(y1,ans_w_f);
soundsc(sou_nf,fs);

%FOR 'IY' sound
ans_w3f=conv(himp3,gRflip);
sou_n3f=conv(y1,ans_w3f);
soundsc(sou_n3f,fs);
flr_song = cat(1,sou_n3,zeros(1000,1),sou_n3f);
soundsc(flr_song);
```

## **Part 5.1**

```
figure, subplot(2,2,1), plot(soun_exp1(1:1000)), title('AA in Exponential');
subplot(2,2,2), plot(soun_exp3(1:1000)), title('IY in Exponential');
subplot(2,2,3), plot(sou_nf(1:1000)), title('AA in Rosenberg flipped');
subplot(2,2,4), plot(sou_n3f(1:1000)), title('IY in Rosenberg flipped');
figure, subplot(2,2,1), plot(sou_n(1:1000)), title('AA in Rosenberg');
subplot(2,2,2), plot(sou_n3(1:1000)), title('IY in Rosenberg');
subplot(2,2,3), plot(sou_nf(1:1000)), title('AA in flipped Rosenberg');
```



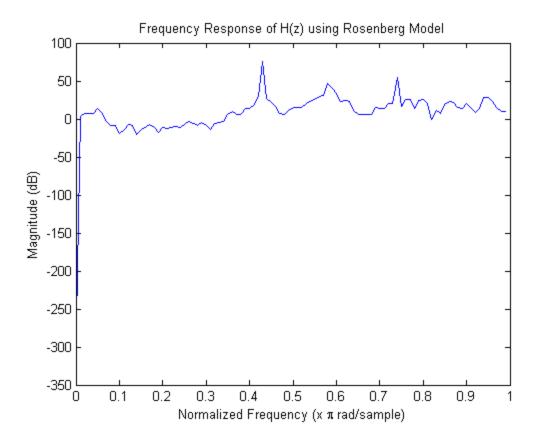


## **Part 5.2**

TO find H(z), first convert all the sytems into Z domain. Multiply V(z), R(z) and G(z) to get H(z). After that plt the frequency response of H(z)

```
syms z n
G_z = symsum((0.5*(1-cos((pi*n)/40))*z^(-n)), n, 0, 40) + symsum(cos((pi*(n-40)/(2
R_z = 1-(1/z);
[r4, D4, G4]=atov(IY,rN2);
V_z = G4/ (D4(1)+D4(2)*(1/z) + D4(3)*(1/z^2)+ D4(4)*(1/z^3)+ D4(5)*(1/z^4)+ D4(6)*
Hz = G_z*V_z*R_z;
[NUME,DENO] = numden(Hz);
num_c = double(coeffs(NUME));
den_c = double(coeffs(DENO));

[mag, freq] = freqz(num_c, den_c, 100);
figure,plot(freq/pi,20*log10(abs(mag)))
xlabel('Normalized Frequency (x \pi rad/sample)');
ylabel('Magnitude (dB)');
title('Frequency Response of H(z) using Rosenberg Model');
```



# **Part 5.3**

#### Listening to the output

```
%Rosenberg model
r_song = cat(1,sou_n,zeros(1000,1),sou_n3);
soundsc(r_song);
%Flipped rosenberg model
flr_song = cat(1,sou_n3,zeros(1000,1),sou_n3f);
soundsc(flr_song);
% The sounds can be distinctly differentiated into "AA" and "IY" vowels.
```

EXERCISE 4.2 Part al Show that TN is equal to corefficent of Z-M in the denominator of 1(2) (m) NN = - KN DK (2) = DK-1 (Z) + NK Z \* DK-1 (Z') -> 100 for k=1, 2, ... N Do (2) =1  $D_1(z) = D_0(z) + A_1(z^{-1})$  $D_1(2) = 1 + N_1 Z^{-1} \Rightarrow [: D_0(z^{-1}) = 0] - 0$ 3 r, is co-efficient of z-1 But D(2) = ( - & dx 2-k  $D_1(2) = 1 - \lambda_1 z^{-1} - 2$ For K=1 From () +(2),  $N_1 = -\lambda_1 - 3$ For K=2,  $D_{2}(z) = D_{1}(z) + N_{2}z^{-2}D_{1}(z^{-1})$  $= 1+N,2^{-1}+N_2 z^{-2} (1+N,2)$  $= 1+ \lambda_1 z^1 + \lambda_2 z^{-2} + \lambda_1 \lambda_2 z^{-1} - 6$ Bon But  $D_2(2) = (-1, 2^{-1} - 1, 2^{-1} - 1, 2^{-1} - 1, 2^{-1})$ From 4 45 => 12 = -62

Similarly,

$$D_{N}(z) = 1 + N_{N-1} z^{-N+1} + N_{N} z^{-N} N_{N-1} N_{N} z^{-N-1}$$

and also

 $D_{N}(z) = 1 - L_{1} z^{-1} - L_{2} z^{-2} + ... - L_{N} z^{-N}$ 
 $D_{N}(z) = 1 - L_{1} z^{-1} - L_{2} z^{-2} + ... - L_{N} z^{-N}$ 
 $D_{N}(z) = 1 - L_{1} z^{-1} - L_{2} z^{-2} + ... - L_{N} z^{-N}$ 
 $D_{N}(z) = 1 - L_{1} z^{-1} - L_{2} z^{-1} + ... - L_{N} z^{-N}$ 
 $D_{N}(z) = 1 - L_{1} z^{-1} - L_{1} z^{-N}$ 
 $D_{N}(z) = 1 - L_{1} z^{-1} - L_{1} z^{-N}$ 
 $D_{N}(z) = 1 - L_{1} z^{-N}$ 
 $D_{N}(z)$ 

: LHS = RHS

$$\int_{K} (z) = \int_{K-1} (z) + \int_{K} z^{-k} \int_{K-1} (z^{-1})$$

$$\int_{k}^{\infty} (z^{2}) = \int_{k-1}^{\infty} (z^{2}) + \int_{k}^{\infty} z^{-k} \int_{k-1}^{\infty} (z^{-1})^{2}$$

$$(Cz) = \int_{k-1}^{2} (Cz) = \int_{k}^{2} z^{-k} D_{k-1} (z^{-1})$$

$$\int_{k}^{2} (z^{-1}) dz = \int_{k}^{2} z^{-k} D_{k-1} (z^{-1})$$

$$P_{K} = \frac{D_{1}(z) - D_{0}(z)}{z^{-1} D_{0}(z^{-1})}$$

$$N_{H} = \frac{D_{N}(z) - D_{N-1}(z)}{z^{-1} D_{N-1}(z^{-1})}$$

$$\lambda_1 = \frac{2.6 - 1.6}{2.6 + 1.6} = \frac{1}{4.2} = 0.23$$

$$N_2 = \frac{A_3 - A_2}{A_1 + A_2} = \frac{0.65 - 2.6}{0.65 + 2.6} = \frac{1.95}{3.25} = 0.6$$

Scanned by CamScanner

Similarly 
$$n_6 = \frac{6.5 - 4}{6.5 + 4} = \frac{2.5}{10.5} = 0.238$$

$$D(z) = 1 - 0.0460 z^{-1} - 0.62322^{-1} + 0.38142^{-3} + 0.24432^{-4} + 0.19732^{-5}$$
  
 $+ 0.26732^{-6} + 0.36552^{-7} = 0.48062^{-8} = 0.11532^{-9} + 0.71z^{-10}$