

Analog final lab assignment

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Experiment 1:

Code:

```
clc
clear
close all
```

Read the audio file and get its sampling frequency

```
[y, F] = audioread('eric.wav');
ty = length(y)/F;
t = linspace(0, ty, length(y));
```

Input signal conversion to frequency domain (to apply ideal LPF)

```
z = fftshift(fft(y));
f = linspace(-F/2, F/2, length(y));
figure
plot(f, abs(z), 'g');
title('Original Signal in Frequency Domain');
```

generate ideal filter

```
% Searching for indices where f = -4000 and f = 4000
for i = 1:length(f)
    if (abs((f(i)+4.0000e+3)) < 0.01)
        index1 = i;
    end
    if (abs((f(i)-4.0000e+3)) < 0.01)
        index2 = i;
        break;
    end
end

% Generating a rect from index1 to index2
range = index2-index1;
step = [zeros(1, index1) ones(1, range) zeros(1, length(f)-index2)];
step = step.';
```

Multiplying input signal by rect to eliminate frequencies other than 4k

```
yfilteredFreq = step.*z;
yfilteredFreqAbs = abs(step.*z);

figure
plot(f, yfilteredFreqAbs, 'g');
xlim([-15000 15000]);
title('Filtered Signal in Frequency Domain');

% Converting the filtered signal to time domain to be modulated
yfiltered = ifft(fftshift(yfilteredFreq));
figure
plot(t, yfiltered, 'r');
title('Filtered Signal in Time Domain');

% Signal after low pass filter
sound(yfiltered, F);
pause(9);
```

Filtered signal resampling

```
maxAmplitude = max(yfiltered);
yfiltered = resample(yfiltered, 500000, F);
yfiltered = yfiltered.';
interval = length(yfiltered);
```

DSB-SC Modulation and Demodulation

Modulation

```
t = linspace(0, ty, interval);
dsbsc = 5*cos(2*pi*100000*t).*yfiltered;
figure
plot(t, dsbsc, 'r');
title('Modulated DSB-SC Signal in Time Domain');
zmsc = abs(fftshift(fft(dsbsc)));
fmsc = linspace(-500000/2, 500000/2, length(dsbsc));
figure
plot(fmsc, zmsc, 'g');
xlim([-150000 150000]);
ylim([0 300]);
title('Modulated DSB-SC Signal in Frequency Domain');
```

Demodulation

```
dsbscEnvelope = abs(hilbert(dsbsc));
dsbscEnvelope = resample(dsbscEnvelope, F, 500000);
t = 0:1/F:length(dsbscEnvelope)/(F);
t = t(1:end-1);
figure
plot(t, dsbscEnvelope, 'r');
title('Demodulated DSB-SC Signal in Time Domain');
zdsb = abs(fftshift(fft(dsbscEnvelope)));
fdsb = linspace(-F/2, F/2, length(dsbscEnvelope));
figure
plot(fdsb, zdsb, 'g');
xlim([-8000 8000]);
ylim([0 1500]);
title('Demodulated DSB-SC Signal in Frequency Domain');
sound(dsbscEnvelope, F);
pause(9);
```

Coherent detection

```
dsbsc0 = awgn(dsbsc, 0);
dsbsc10 = awgn(dsbsc, 10);
dsbsc30 = awgn(dsbsc, 30);
t = linspace(0, ty, length(dsbsc));
vpe = dsbsc.*cos((2*pi*100000*t) + deg2rad(20));
vfe = dsbsc.*cos(2*pi*100100*t);
v0 = dsbsc0.*cos(2*pi*100000*t);
v10 = dsbsc10.*cos(2*pi*100000*t);
v30 = dsbsc30.*cos(2*pi*100000*t);
v0 = resample(v0, F, 500000);
v10 = resample(v10, F, 500000);
v30 = resample(v30, F, 500000);
vpe = resample(vpe, F, 500000);
vfe = resample(vfe, F, 500000);
zsc0 = fftshift(fft(v0));
zsc10 = fftshift(fft(v10));
zsc30 = fftshift(fft(v30));
zscpe = fftshift(fft(vpe));
zscfe = fftshift(fft(vfe));
fsc = linspace(-F/2, F/2, length(v0));
for i = 1:length(fsc)
    if (abs((fsc(i)+4.0000e+3)) < 0.01)
        index1 = i;
    end
    if (abs((fsc(i)-4.0000e+3)) < 0.01)
        index2 = i;
        break;
    end
end
end
```

```

range = index2-index1;
step = [zeros(1, index1) ones(1, range) zeros(1, length(fsc)-(index2))];
dsbscCoherent0 = step.*zsc0;
dsbscCoherent10 = step.*zsc10;
dsbscCoherent30 = step.*zsc30;
dsbscCoherenttpe = step.*zscpe;
dsbscCoherenttfe = step.*zscfe;
figure
plot(fsc, abs(dsbscCoherent0), 'g');
title('Demodulated DSB-SC Signal using Coherent Detection with SNR = 0 in Frequency Domain');
xlim([-8000 8000]);
figure
plot(fsc, abs(dsbscCoherent10), 'g');
title('Demodulated DSB-SC Signal using Coherent Detection with SNR = 10 in Frequency Domain');
xlim([-8000 8000]);
figure
plot(fsc, abs(dsbscCoherent30), 'g');
title('Demodulated DSB-SC Signal using Coherent Detection with SNR = 30 in Frequency Domain');
xlim([-8000 8000]);
figure
plot(fsc, abs(dsbscCoherenttpe), 'g');
title('Demodulated DSB-SC Signal using Coherent Detection with Phase Error = 20 in Frequency Domain');
xlim([-8000 8000]);
figure
plot(fsc, abs(dsbscCoherenttfe), 'g');
title('Demodulated DSB-SC Signal using Coherent Detection with Frequency Error = 100 HZ in Frequency Domain');
xlim([-8000 8000]);
dsbscCoherentTime0 = ifft(fftshift(dsbscCoherent0));
dsbscCoherentTime10 = ifft(fftshift(dsbscCoherent10));
dsbscCoherentTime30 = ifft(fftshift(dsbscCoherent30));
dsbscCoherentTimepe = ifft(fftshift(dsbscCoherenttpe));
dsbscCoherentTimefe = ifft(fftshift(dsbscCoherenttfe));
t = linspace(0, ty, length(dsbscCoherentTime0));
figure
plot(t, abs(dsbscCoherentTime0), 'r');
title('Demodulated DSB-SC Signal using Coherent Detection with SNR = 0 in Time Domain');
sound(abs(dsbscCoherentTime0), F);
pause(9);
figure
plot(t, abs(dsbscCoherentTime10), 'r');
title('Demodulated DSB-SC Signal using Coherent Detection with SNR = 10 in Time Domain');
sound(abs(dsbscCoherentTime10), F);

```

```
pause(9);
figure
plot(t, abs(dsbscCoherentTime30), 'r');
title('Demodulated DSB-SC Signal using Coherent Detection with SNR = 30 in Time Domain');
sound(abs(dsbscCoherentTime30), F);
pause(9);
figure
plot(t, abs(dsbscCoherentTimepe), 'r');
title('Demodulated DSB-SC Signal using Coherent Detection with Phase Error = 20 in Time Domain');
sound(abs(dsbscCoherentTimepe), F);
pause(9);
figure
plot(t, abs(dsbscCoherentTimefe), 'g');
title('Demodulated DSB-SC Signal using Coherent Detection with Frequency Error = 100 HZ in Time Domain');
sound(abs(dsbscCoherentTimefe), F);
pause(9);
```

DSB-TC Modulation and Demodulation

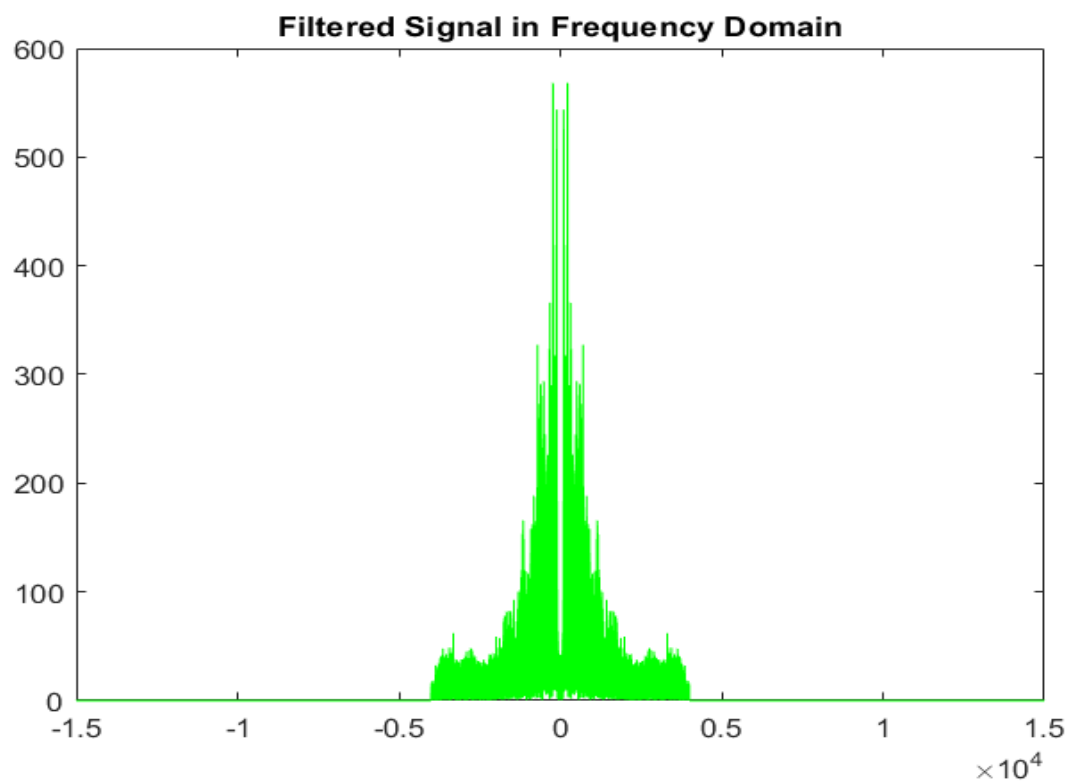
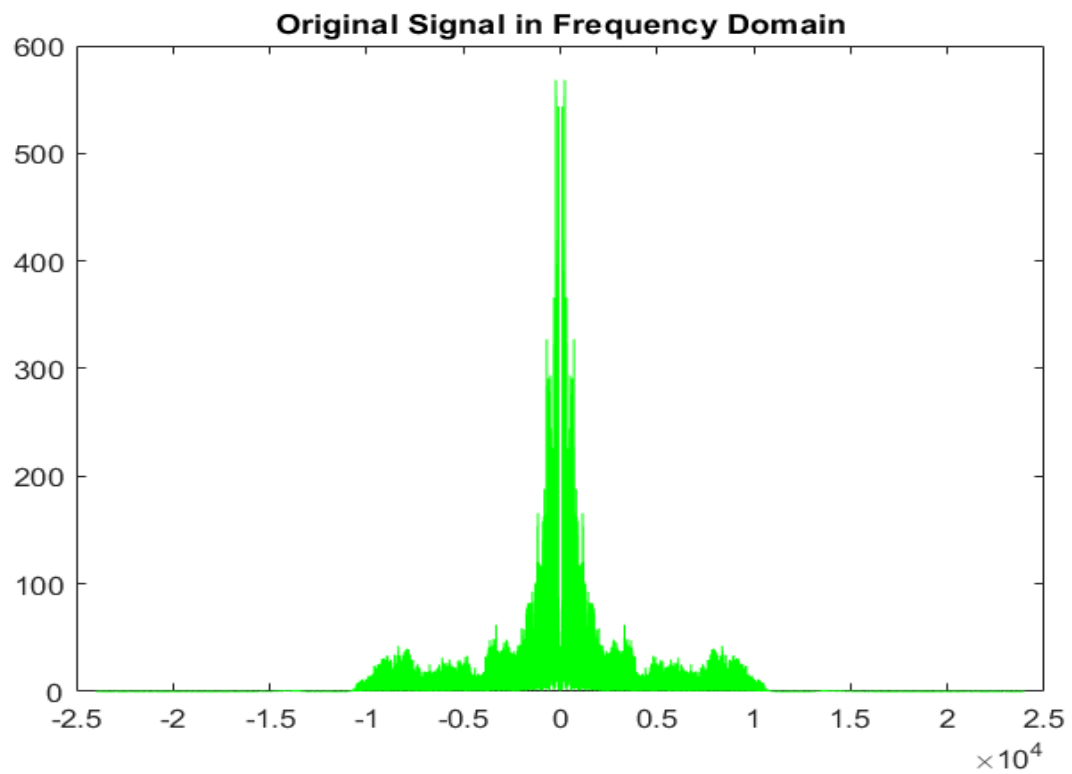
Modulation

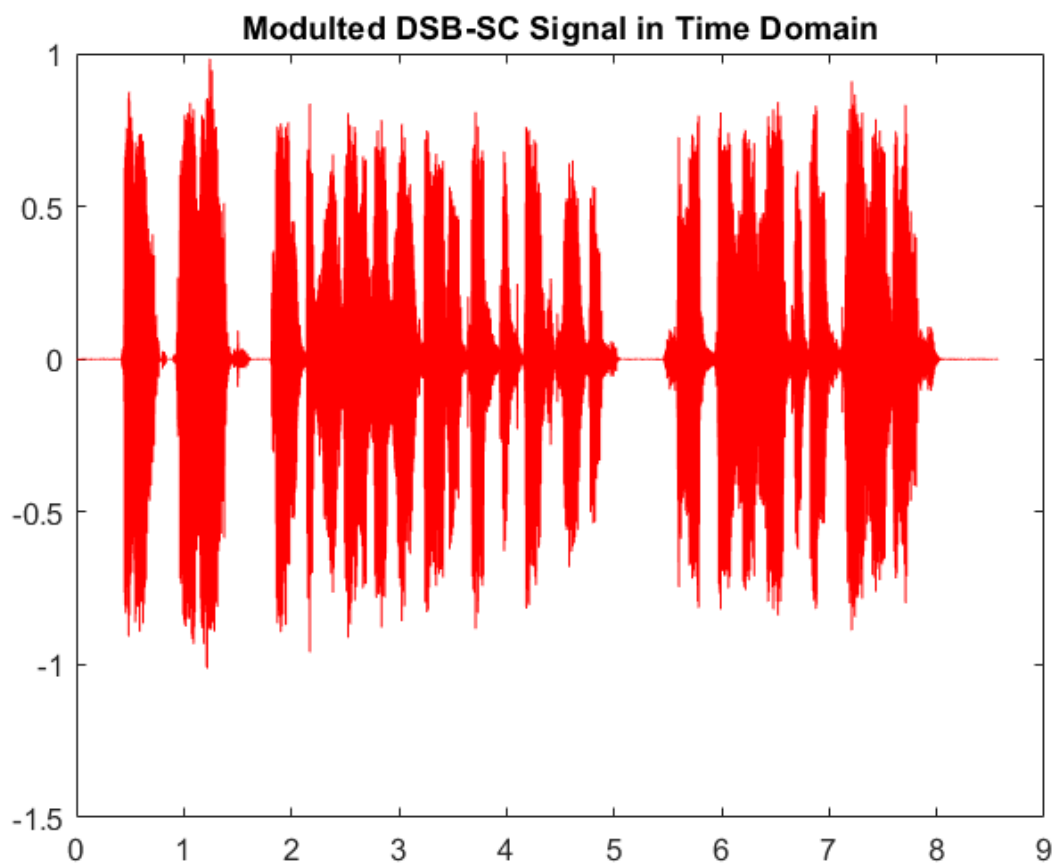
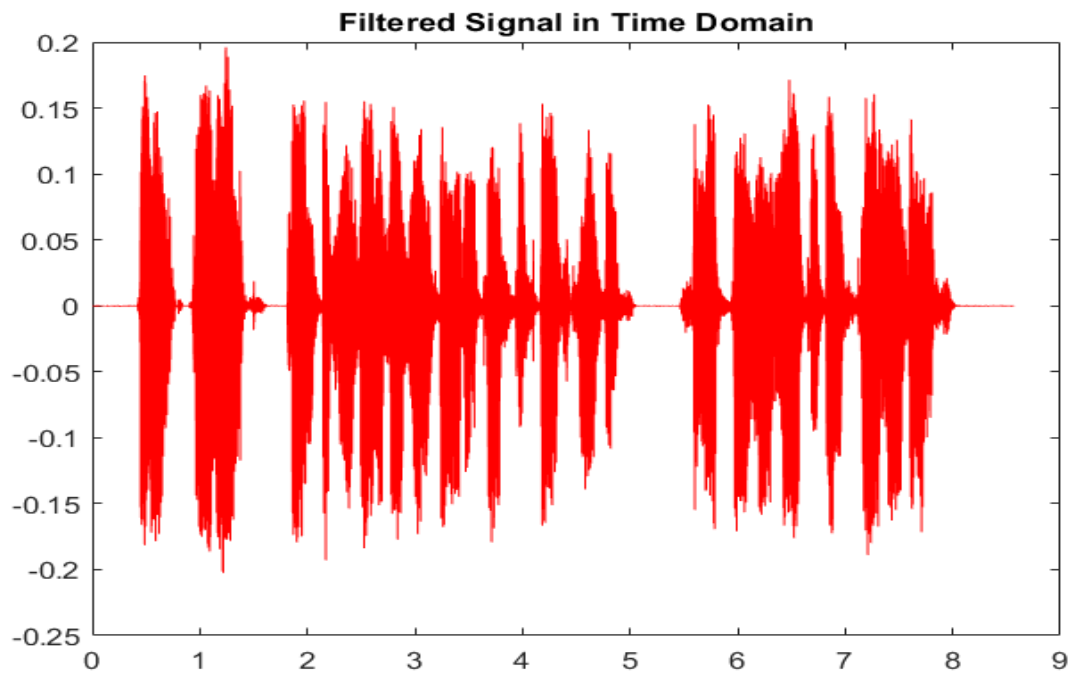
```
t = linspace(0, ty, interval);
dsbtc =
2.*maxAmplitude.*(1+(0.5/maxAmplitude).*yfiltered).*cos(2.*pi.*100000.*t);
figure
plot(t, dsbtc, 'r');
title('Modulated DSB-TC Signal in Time Domain');
zmtc = abs(fftshift(fft(dsbtc)));
fmtc = linspace(-500000/2, 500000/2, length(dsbtc));
figure
plot(fmtc, zmtc, 'g');
xlim([-150000 150000]);
ylim([0 1000]);
title('Modulated DSB-TC Signal in Frequency Domain');
```

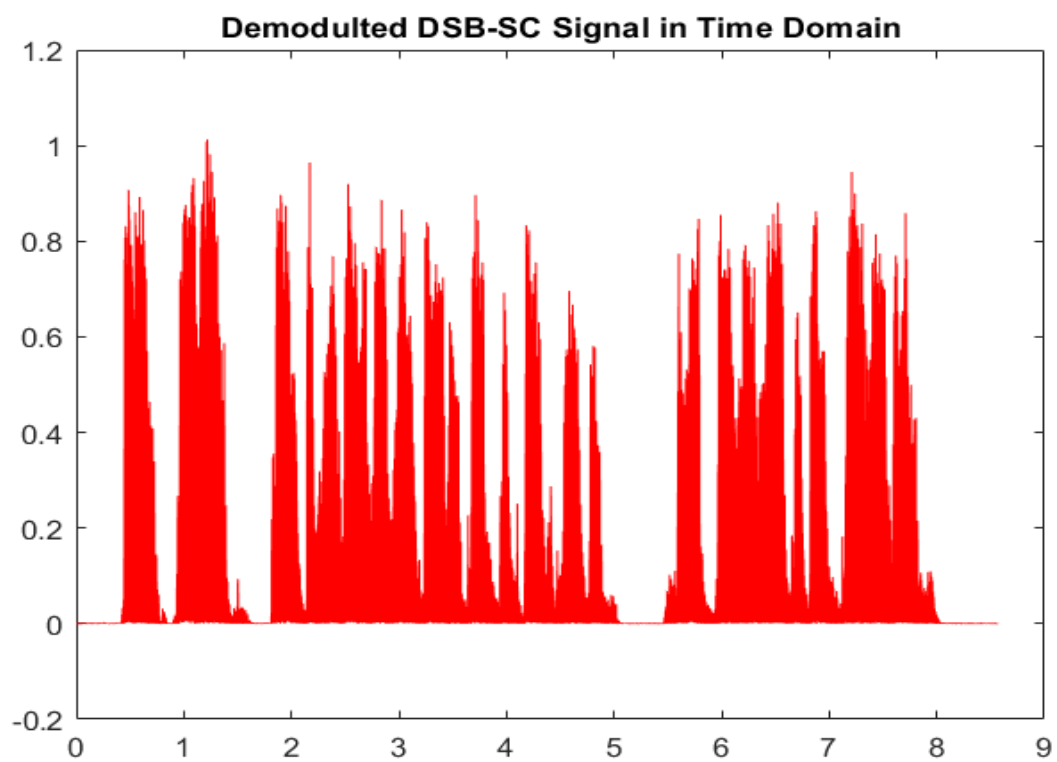
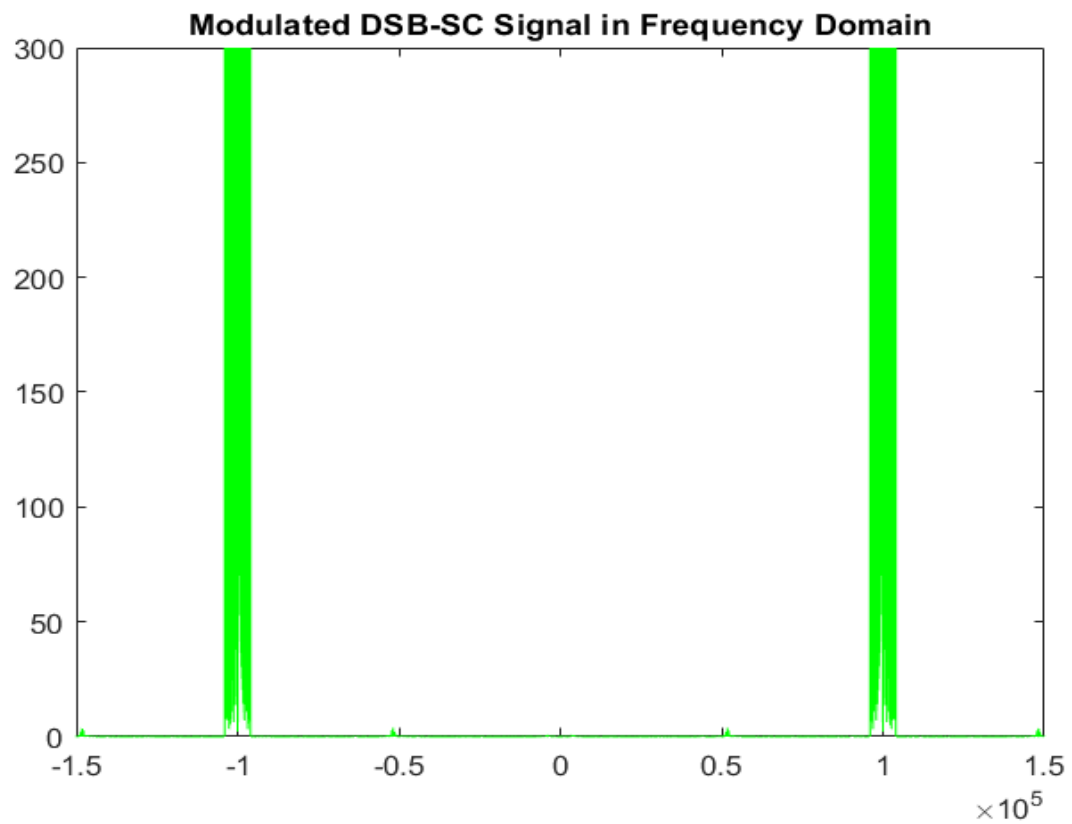
Demodulation

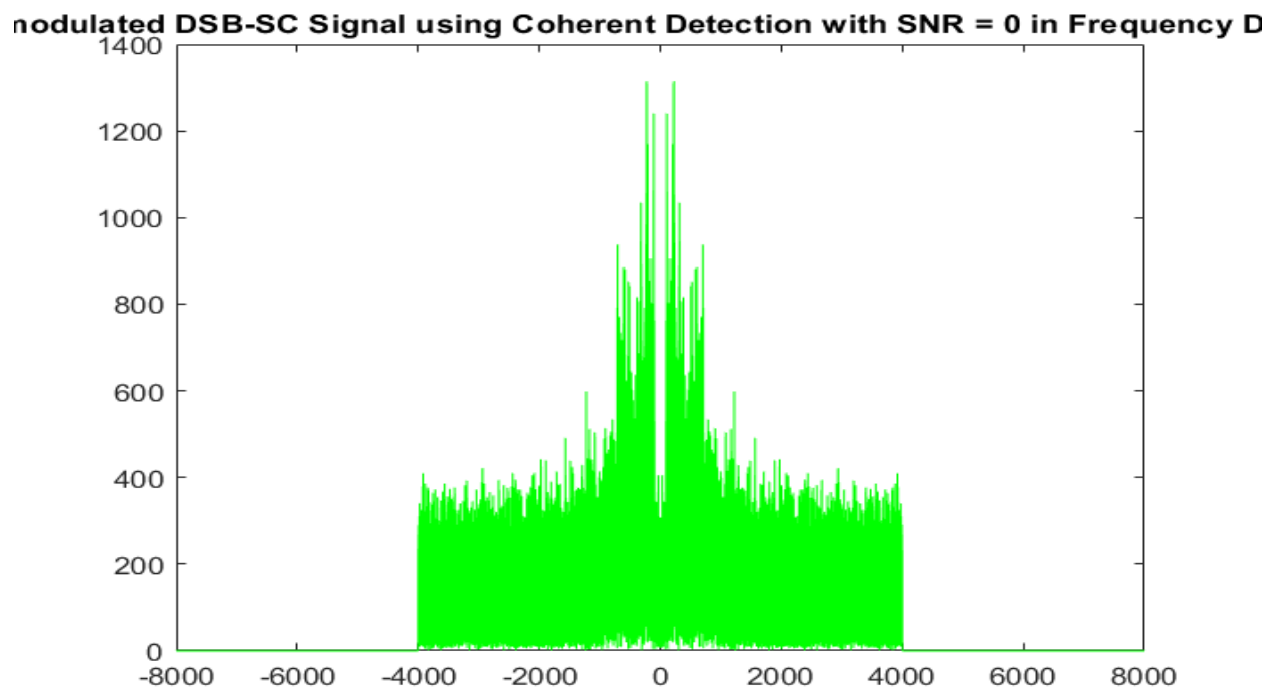
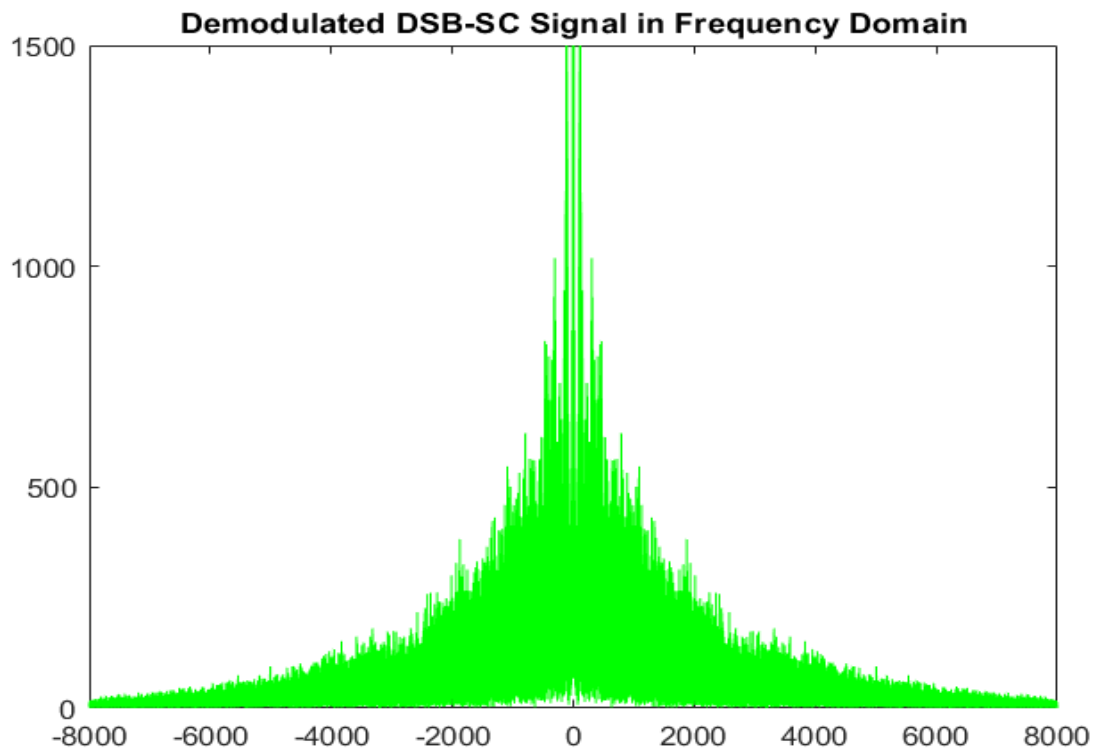
```
dsbtcEnvelope = abs(hilbert(dsbtc));
dsbtcEnvelope = resample(dsbtcEnvelope, F, 500000);
dsbtcEnvelope=dsbtcEnvelope-mean(dsbtcEnvelope);
t = 0:1/F:length(dsbtcEnvelope)/(F);
t = t(1:end-1);
figure
plot(t, dsbtcEnvelope, 'r');
title('Demodulated DSB-TC Signal in Time Domain');
zdtc = abs(fftshift(fft(dsbtcEnvelope)));
fdtc = linspace(-F/2, F/2, length(dsbtcEnvelope));
figure
plot(fdtc, zdtc, 'g');
xlim([-8000 8000]);
ylim([0 1000]);
title('Demodulated DSB-TC Signal in Frequency Domain');
sound(dsbtcEnvelope, F);
```

output:

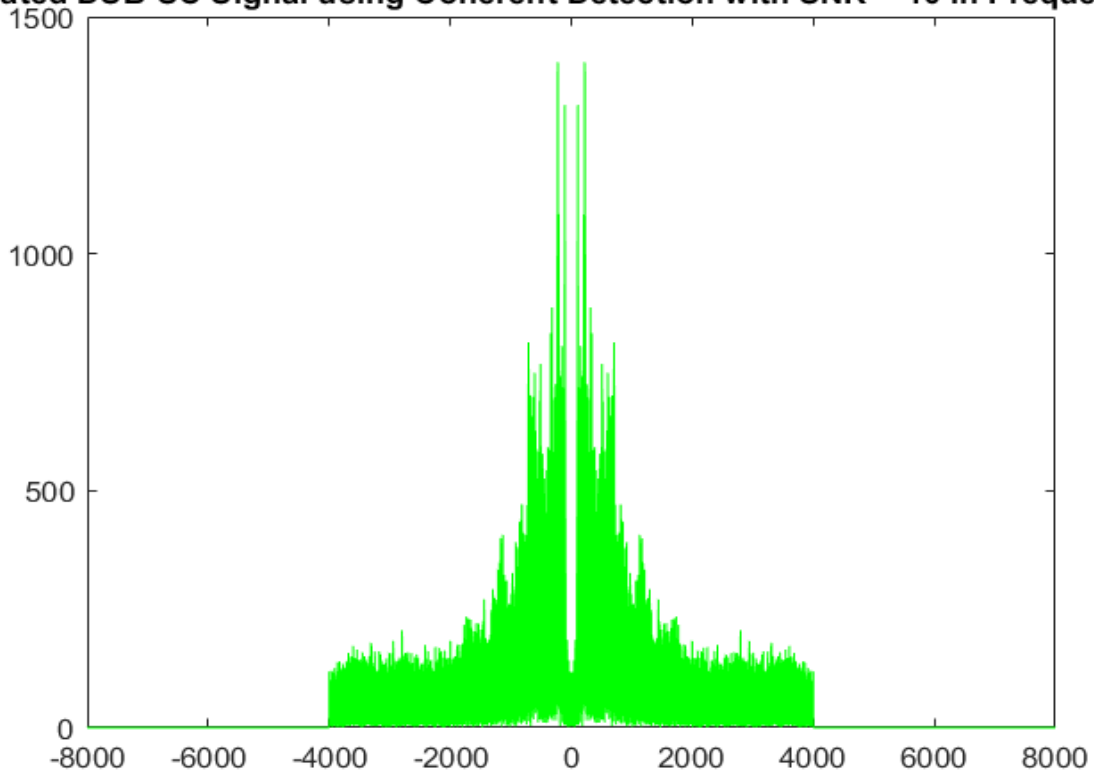




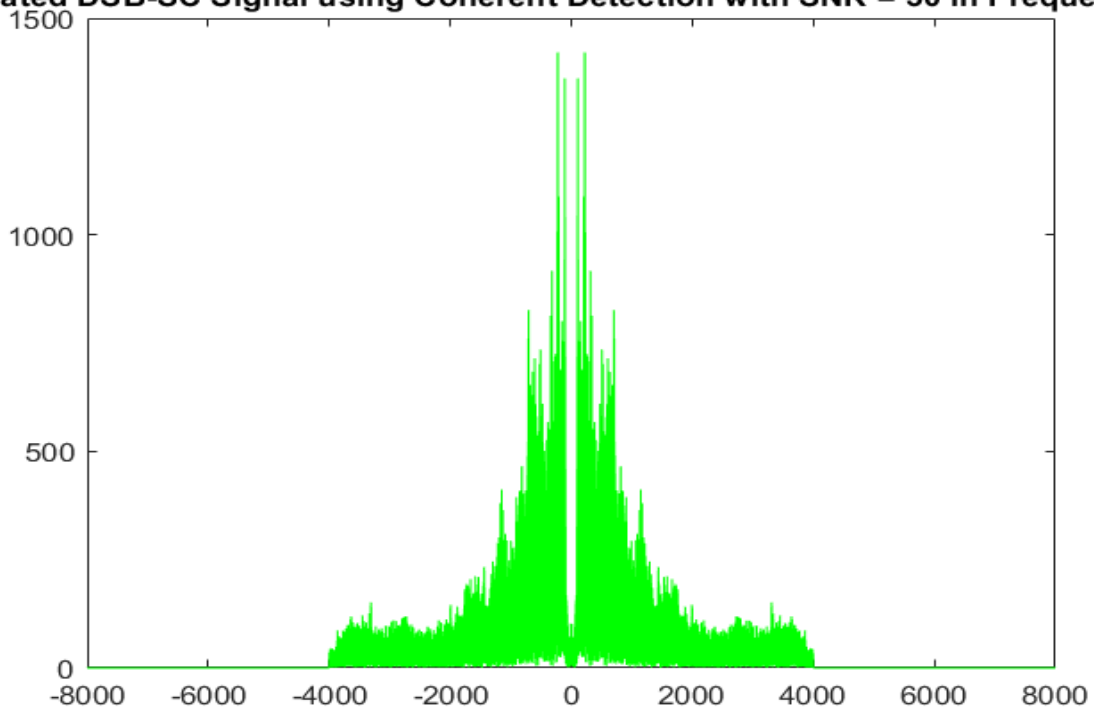




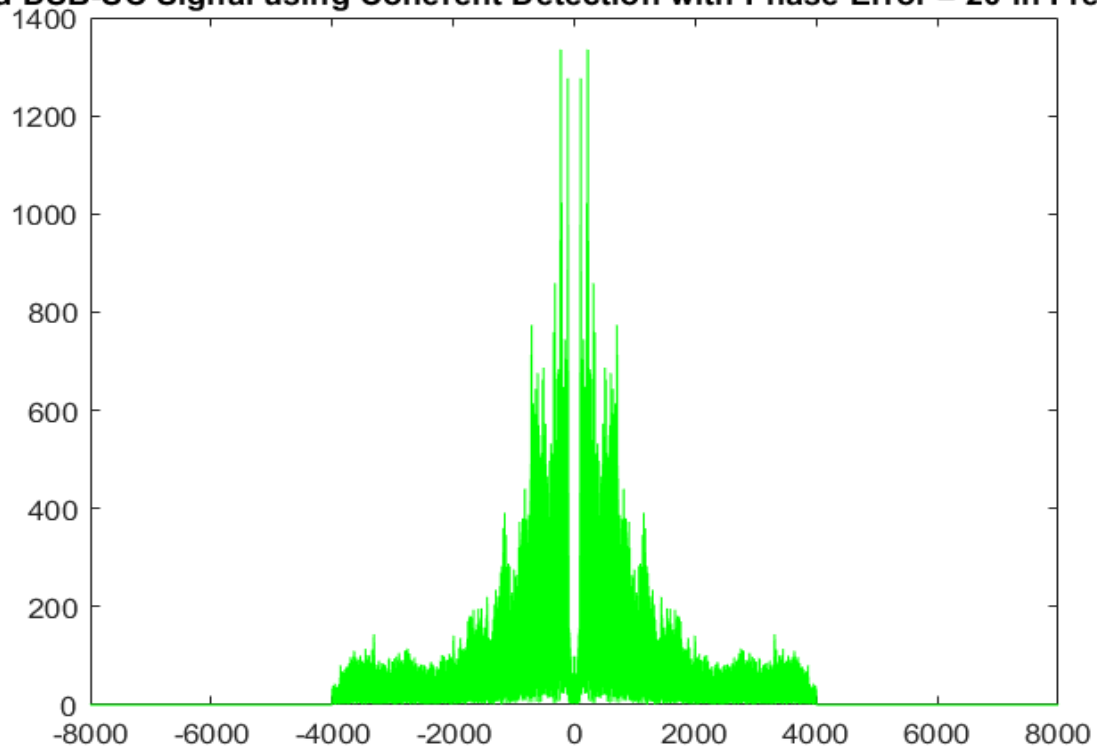
Modulated DSB-SC Signal using Coherent Detection with SNR = 10 in Frequency I



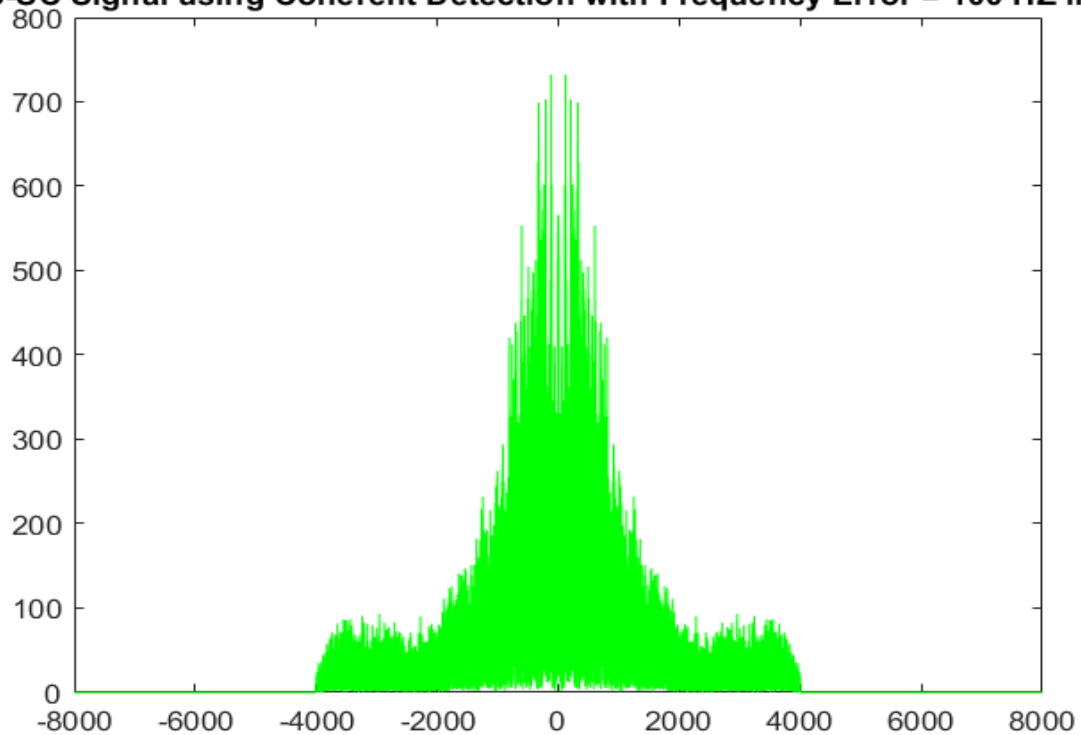
Modulated DSB-SC Signal using Coherent Detection with SNR = 30 in Frequency I



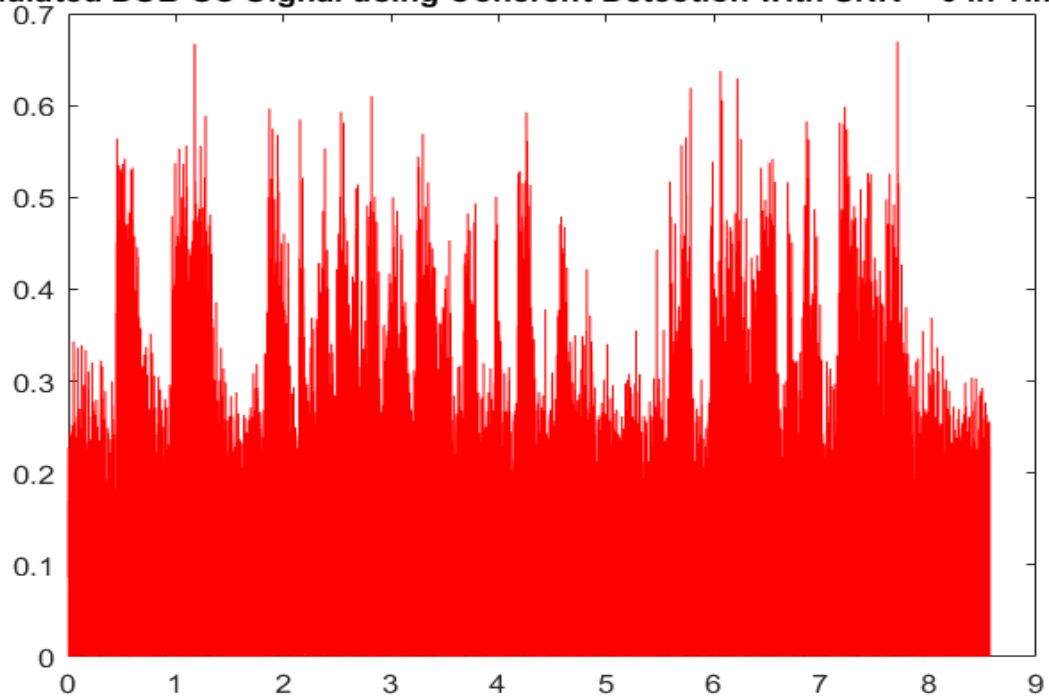
ulated DSB-SC Signal using Coherent Detection with Phase Error = 20 in Frequen



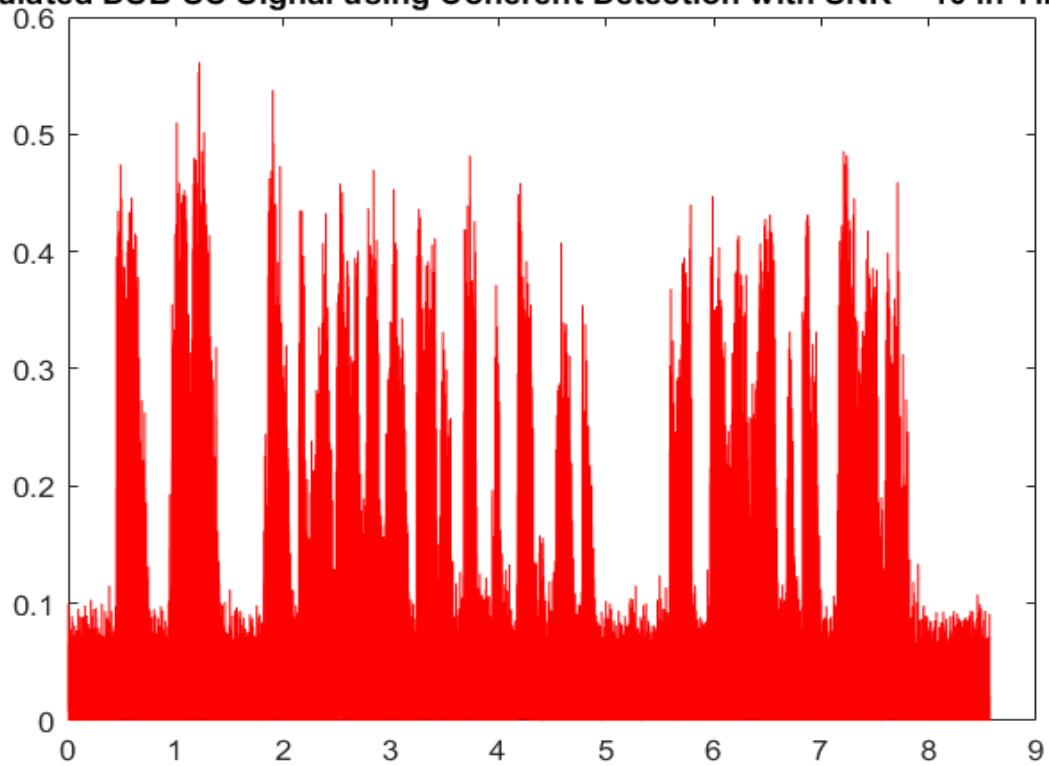
d DSB-SC Signal using Coherent Detection with Frequency Error = 100 HZ in Fre



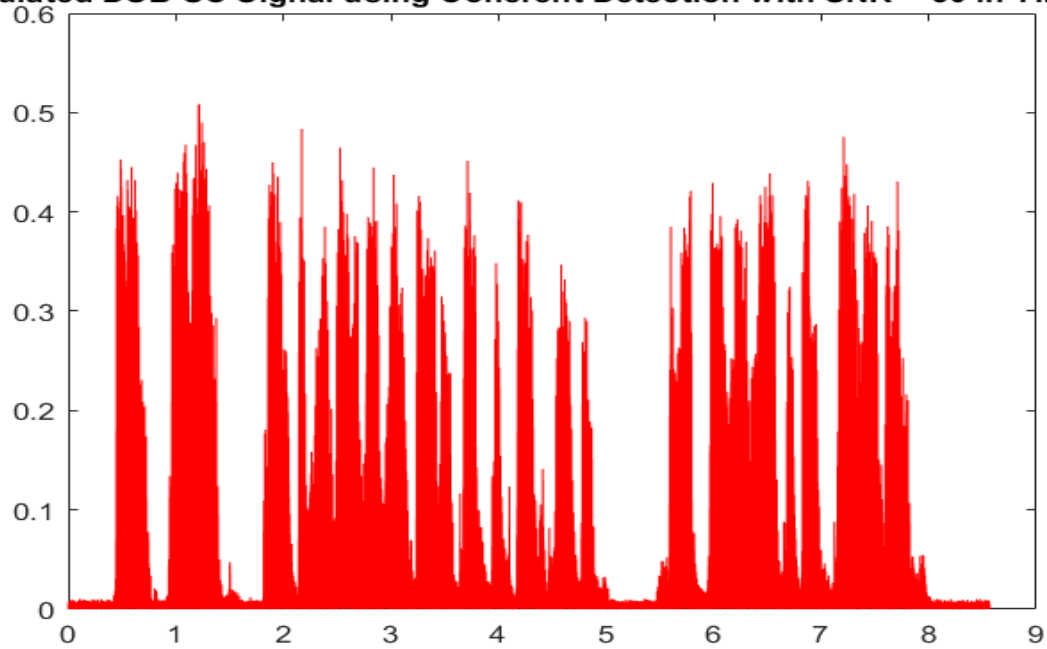
Demodulated DSB-SC Signal using Coherent Detection with SNR = 0 in Time Domain



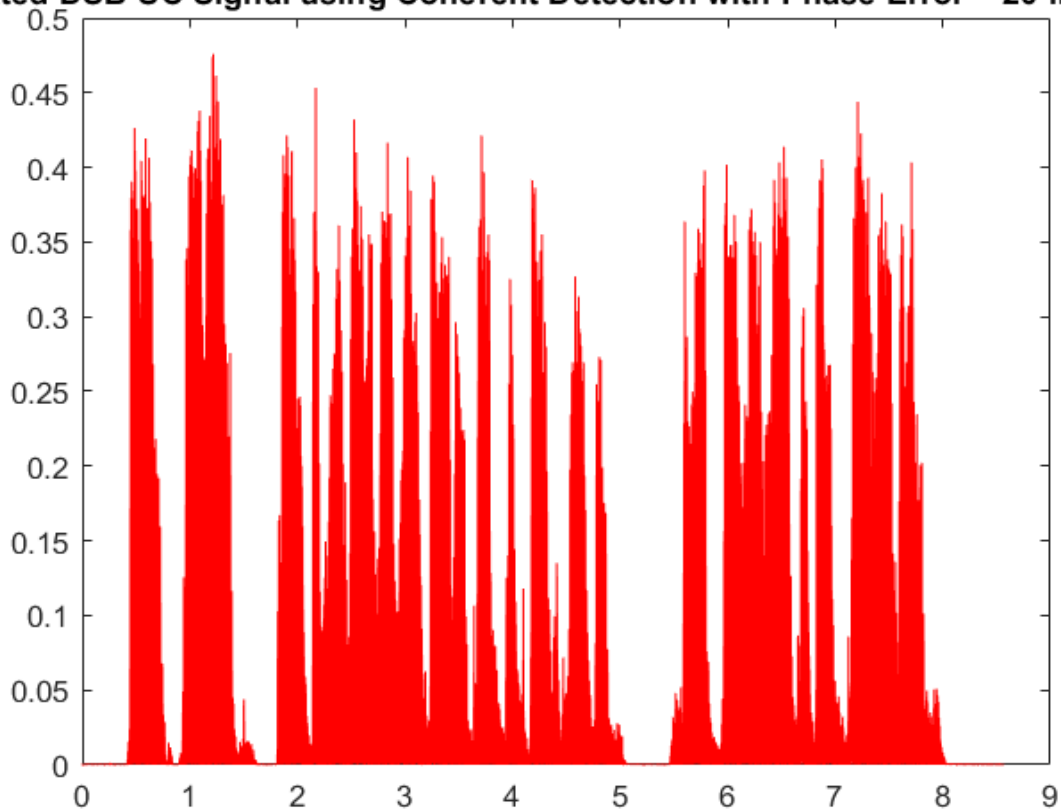
Demodulated DSB-SC Signal using Coherent Detection with SNR = 10 in Time Domain



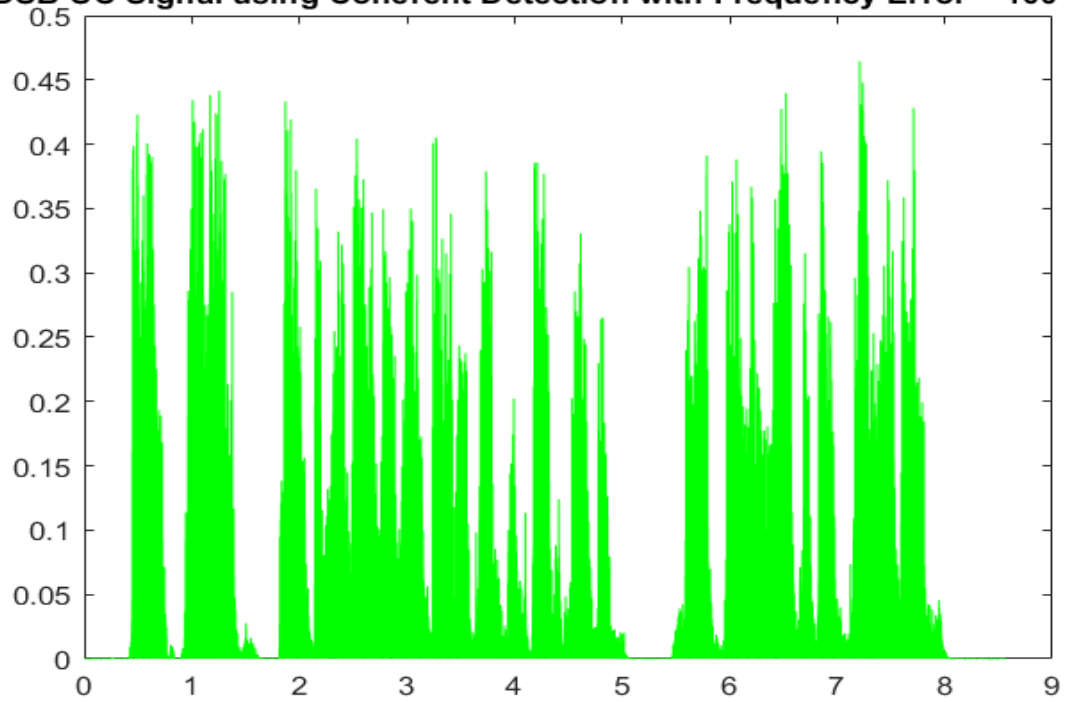
Demodulated DSB-SC Signal using Coherent Detection with SNR = 30 in Time Domain



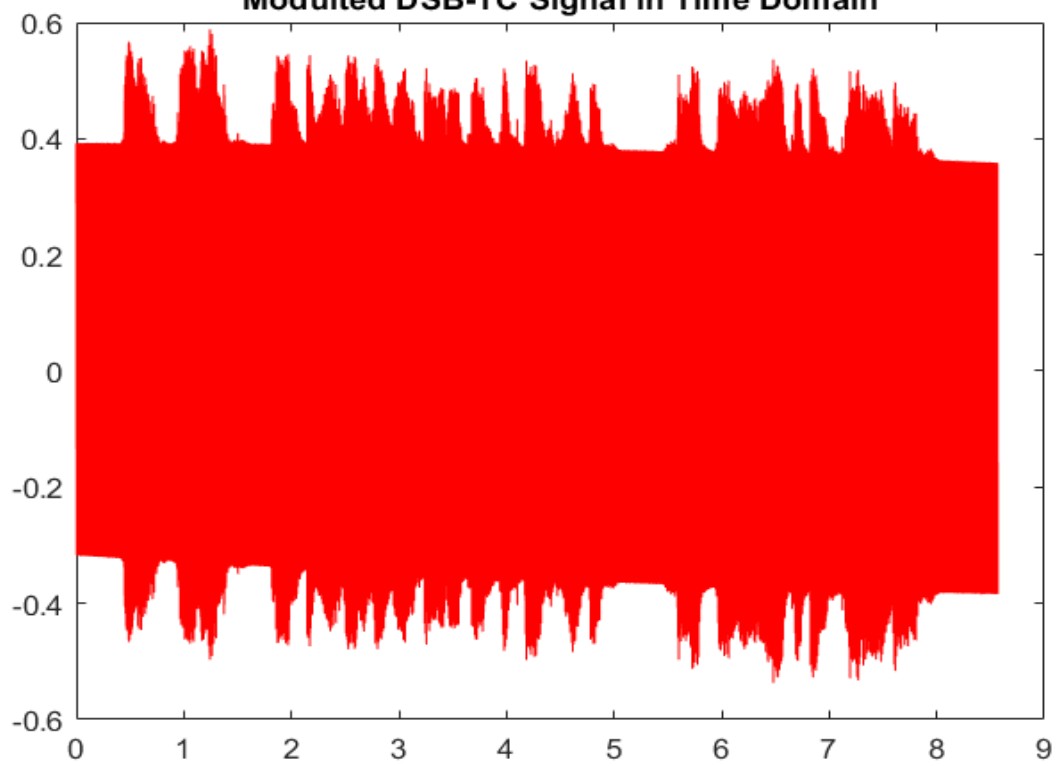
Demodulated DSB-SC Signal using Coherent Detection with Phase Error = 20 in Time Domain

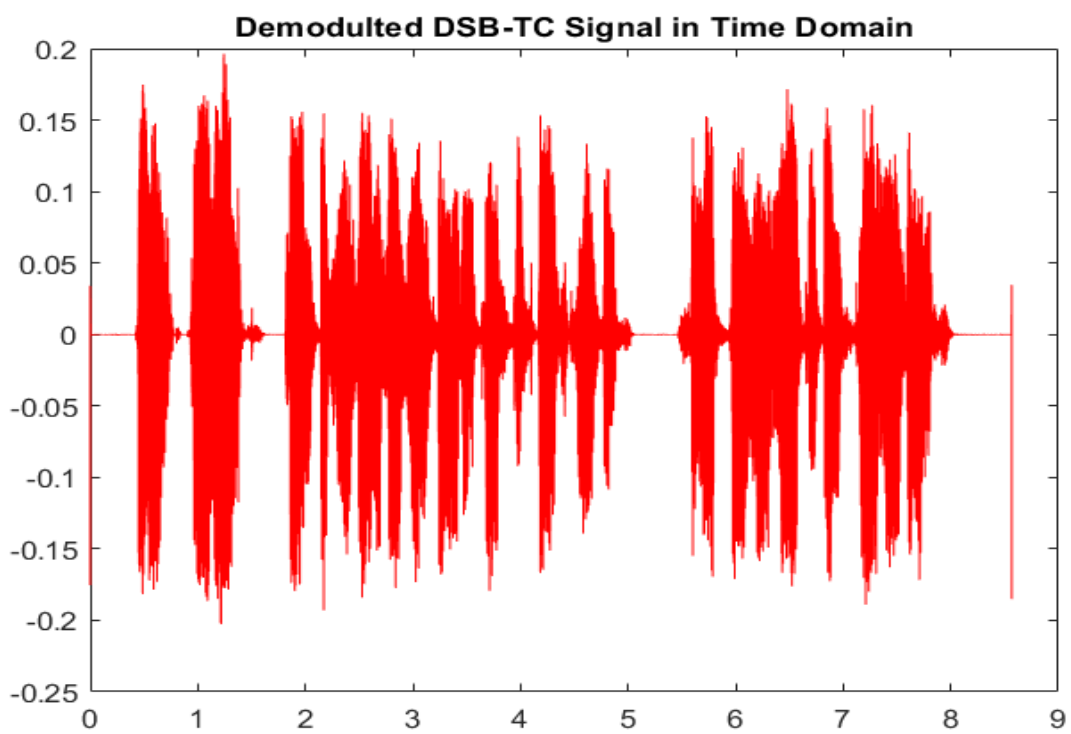
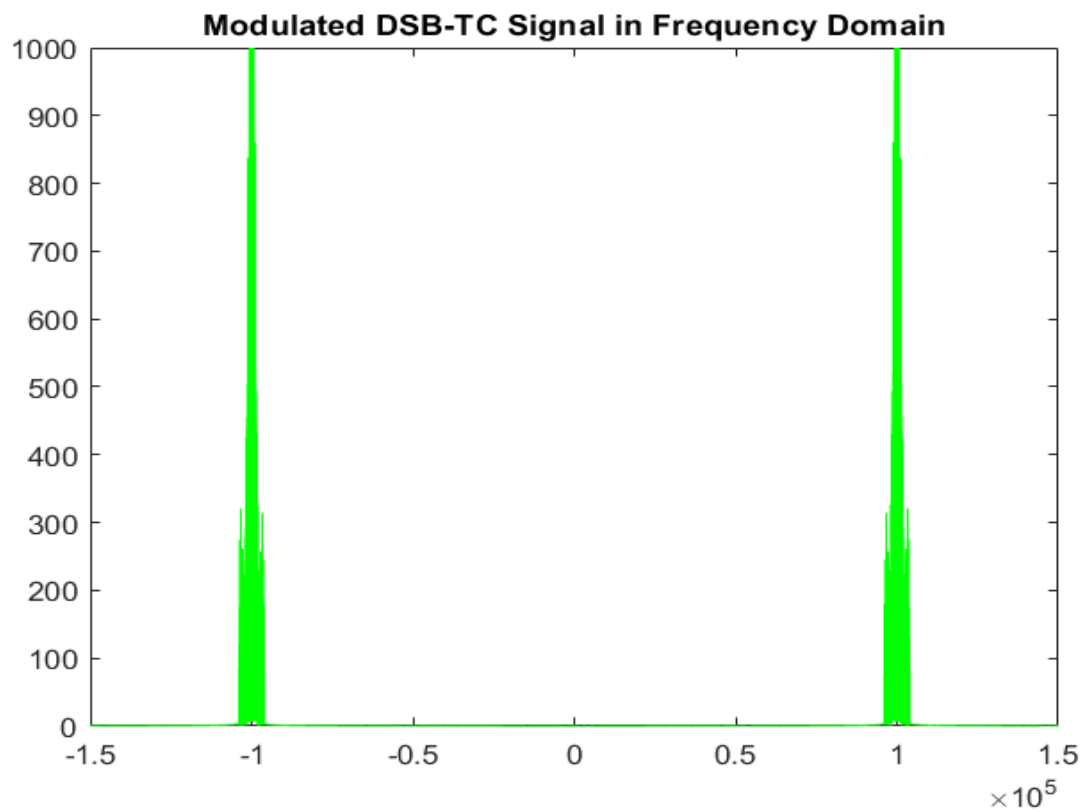


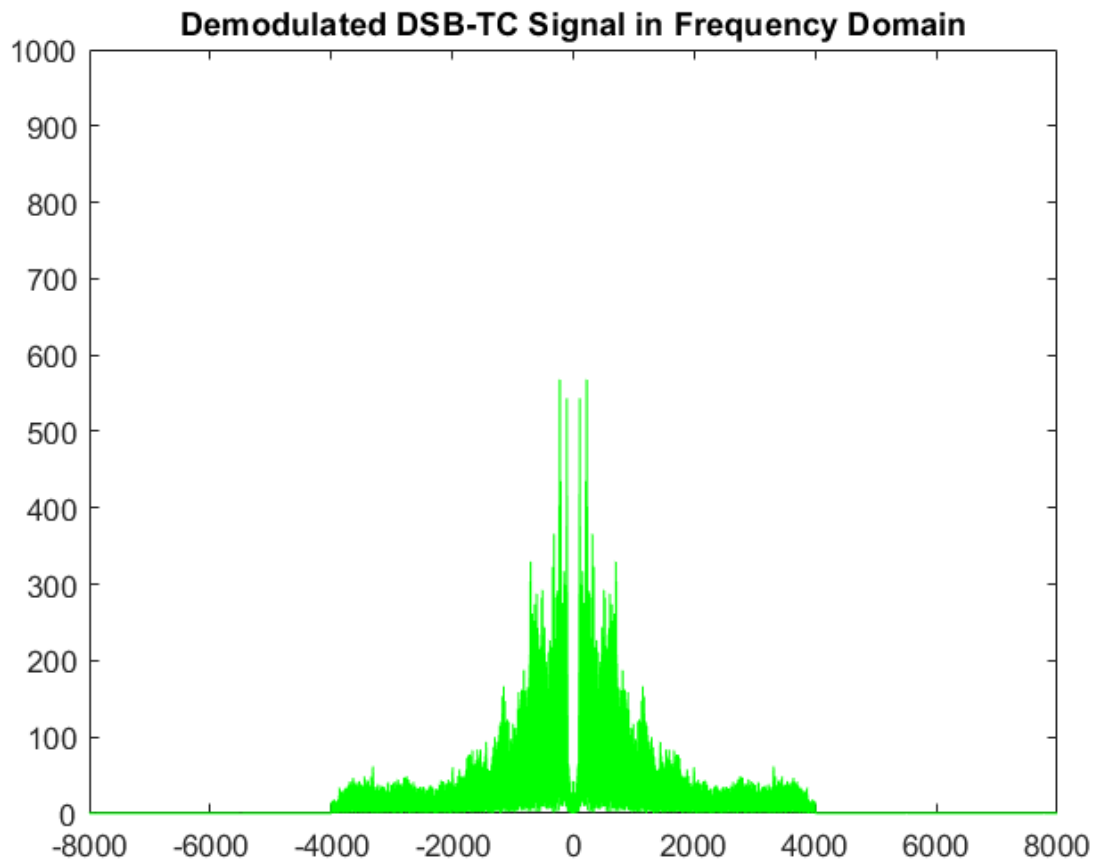
ated DSB-SC Signal using Coherent Detection with Frequency Error = 100 HZ in 1



Modulated DSB-TC Signal in Time Domain







Experiment Conclusions

→ By using Envelope Detector, we observe that there is more distortion in case of DSB-SC. DSB-SC should be demodulated by coherent detection. Envelope Detector can be used to demodulate DSB-TC.

→ It is called beat effect

Experiment 2:

Code:

```
clc
clear
close all
```

Read the audio file and get its sampling frequency

```
[y, F] = audioread('eric.wav');
ty = length(y)/F;
t = linspace(0, ty, length(y));
T=linspace(0, ty, length(y));
plot(t,y,'r');
title('Original Signal in Time Domain');
```

Input signal conversion to frequency domain

```
z = fftshift(fft(y));
zf=abs(z);
f = linspace(-F/2, F/2, length(y));
plot(f,zf,'g');
title('Original Signal in Frequency Domain');
```

apply ideal LPF ,Searching for indices where $f = -4000$ and $f = 4000$

```
for i = 1:length(f)
    if (abs((f(i)+4.0000e+3)) < 0.01)
        index1 = i;
    end
    if (abs((f(i)-4.0000e+3)) < 0.01)
        index2 = i;
        break;
    end
end

%Generating a rect from index1 to index2
range = index2-index1;
step = [zeros(1, index1) ones(1, range) zeros(1, length(f)-index2)];
step = step.';
```

Multiplying input signal by rect to eliminate frequencies other than 4k

```
yfilteredFreq = step.*z;  
yfilteredFreqAbs = abs(step.*z);
```

```
figure  
plot(f, yfilteredFreqAbs, 'g');  
xlim([-4.5 4.5].*10^3)  
title('Filtered Signal in Frequency Domain at 4KHZ (ideal filter)');
```

Converting the filtered signal to time domain to be modulated DSB-SC

```
yfiltered= ifft(fftshift(yfilteredFreq));  
yfiltered=yfiltered.';  
figure  
plot(T,yfiltered, 'r');  
title('Filtered Signal in Time Domain at 4KHZ (ideal filter)');  
sound(yfiltered, F);  
pause(9);
```

```
%Filtered signal resampling  
maxAmplitude = max(yfiltered);  
yfiltered = resample(yfiltered, 500000, F);  
interval = length(yfiltered);
```

DSB-SC Modulation

```
t = linspace(0, ty, interval);  
carrier=cos(2*pi*100000*t);  
dsbsc =carrier.*yfiltered;  
figure  
plot(t, dsbsc, 'r');  
title('Modulated DSB-SC Signal in Time Domain');  
zm_sc=fftshift(fft(dsbsc));  
zm_sc = abs(fftshift(fft(dsbsc)));  
fm_sc = linspace(-500000/2, 500000/2, length(dsbsc));  
figure  
plot(fm_sc, zm_sc, 'g');  
xlim([-150000 150000]);  
ylim([0 1000]);  
title('Modulated DSB-SC Signal in Frequency Domain');  
xlim([-12 12].*10^4)
```

remove USB to get SSB using ideal filter

```
for i1 = 1:length(dsb_sc)
    if (abs((f_msc(i1)+1.0000e+5)) < 0.1)
        idx1 = i1;
    end
    if (abs((f_msc(i1)-1.0000e+5)) < 0.1)
        idx2 = i1;
        break;
    end
end

%Generating a rect from idx1 to idx2
range1 = idx2-idx1;
step1 = [zeros(1, idx1) ones(1, range1) zeros(1, length(f_msc)-idx2)];
LSB_Frequency = step1.*z_m_sc;
LSB_FAbs = abs(step1.*z_m_sc);
figure
plot(f_msc, LSB_FAbs, 'g');
xlim([-150000 150000]);
ylim([0 1000]);
title('Obtain LSB in Frequency Domain using from DSB-SC (ideal filter)');
xlim([-12 12].*10^4)
LSB_time = ifft(ifftshift(LSB_Frequency));
```

demodulation of SSB using coherent detector ideal filter

```
SSB_SC = LSB_time.*carrier;
SSB_SC = resample(SSB_SC,F,500000);
SSB_SC_ff= fftshift(fft(SSB_SC)) ;

%ideal lpf to get signal in frequency domain
SSB_SC_ff=SSB_SC_ff(1:end-1);
SSB_SC_ff =SSB_SC_ff.*step';
SSB_SC_time =ifft(ifftshift(SSB_SC_ff)) ;
signal_frequency_domain= fftshift(fft(SSB_SC_time));
figure
plot(T,SSB_SC_time, 'r');
title('recieved LSB in Time Domain (ideal filter)');
figure;
plot(f,abs(signal_frequency_domain), 'g');
title('recieved LSB in Frequency Domain (ideal filter)');
xlim([-4.5 4.5].*10^3)
sound(SSB_SC_time,F);
pause(9);
```

remove USB to get SSB using butter filter

Butterworth filter BPF to get LSB

```
fnorm = 500000/2;
BW_fitter=(500000*4000)/48000;
[numerator, denominator] = butter(4,[100000 100000+BW_fitter]/fnorm,'bandpass');
Filter_DSB = filter(numerator, denominator, dsbsc);
LSB.Butter = Filter_DSB.*carrier;
plot(fmsc,abs(fftshift(fft(Filter_DSB))),'g');
title('Obtain LSB in Frequency Domain (Butter filter)');

%down sample butter filter
LSB_down =resample(LSB.Butter,F,500000);
LSB_down =LSB_down(1:end-1);
lsb_freq=abs(fftshift(fft(LSB_down)));
```

Butterworth filter LPF to get LSB after demodulation

```
[numerator, denominator] = butter(4,4000/(F/2));
LSB_LPF_Time = filter(numerator, denominator, LSB_down);
LSB_LPF_Freq = fftshift(fft(LSB_LPF_Time));
figure
plot(LSB_LPF_Time,'r');
title('recieved LSB in Time Domain (butter filter)');

figure
plot(f,abs(LSB_LPF_Freq),'g');
title('recieved LSB in Frequency Domain (butter filter)');
xlim([-4.5 4.5].*10^3)

%NOISE is added to signal
noised_signal_0 = awgn(LSB_time,0, 'measured');
noised_signal_10 = awgn(LSB_time,10,'measured');
noised_signal_30 = awgn(LSB_time,30,'measured');
noised_signal_0=noised_signal_0';
noised_signal_10=noised_signal_10';
noised_signal_30=noised_signal_30';
carrier=carrier.';
LSB0 = noised_signal_0.*carrier;
coherent0 = resample(LSB0,F,500000);
coherent0 = coherent0(1:end-1);
coherentFreq0 = fftshift(fft(coherent0));
coherentFilter0 = step.*coherentFreq0;
coherentTime0 = ifft(ifftshift(coherentFilter0)) ;
```

```

figure;
plot(T,coherentTime0,'r');
title('recieved LSB in Time Domain (ideal filter) with noise SNR=0');
figure;
plot(f,abs(coherentFilter0),'g');
title('recieved LSB in Freq Domain (ideal filter) with noise SNR=0');
xlim([-4.5 4.5].*10^3)
sound(real(coherentTime0),F);
pause(9);
LSB10 = noised_signal_10.*carrier;
coherent10 = resample(LSB10,F,500000);
coherent10 = coherent10(1:end-1);
coherentFreq10 = fftshift(fft(coherent10));
coherentFilter10 = step.*coherentFreq10;
coherentTime10 = ifft(ifftshift(coherentFilter10)) ;
figure;
plot(T,coherentTime10,'r');
title('recieved LSB in Time Domain (ideal filter) with noise SNR=10');
figure;
plot(f,abs(coherentFilter10),'g');
title('recieved LSB in Freq Domain (ideal filter) with noise SNR=10');
xlim([-4.5 4.5].*10^3)
sound(real(coherentTime10),F);
pause(9);
LSB30 = noised_signal_30.*carrier;
coherent30 = resample(LSB30,F,500000);
coherent30 = coherent30(1:end-1);
coherentFreq30 = fftshift(fft(coherent30));
coherentFilter30 = step.*coherentFreq30;
coherentTime30 = ifft(ifftshift(coherentFilter30)) ;
figure;
plot(T,coherentTime30,'r');
title('recieved LSB in Time Domain (ideal filter) with noise SNR=30');
figure;
plot(f,abs(coherentFilter30),'g');
title('recieved LSB in Freq Domain (ideal filter) with noise SNR=30');
xlim([-4.5 4.5].*10^3)
sound(real(coherentTime30),F);
pause(9);

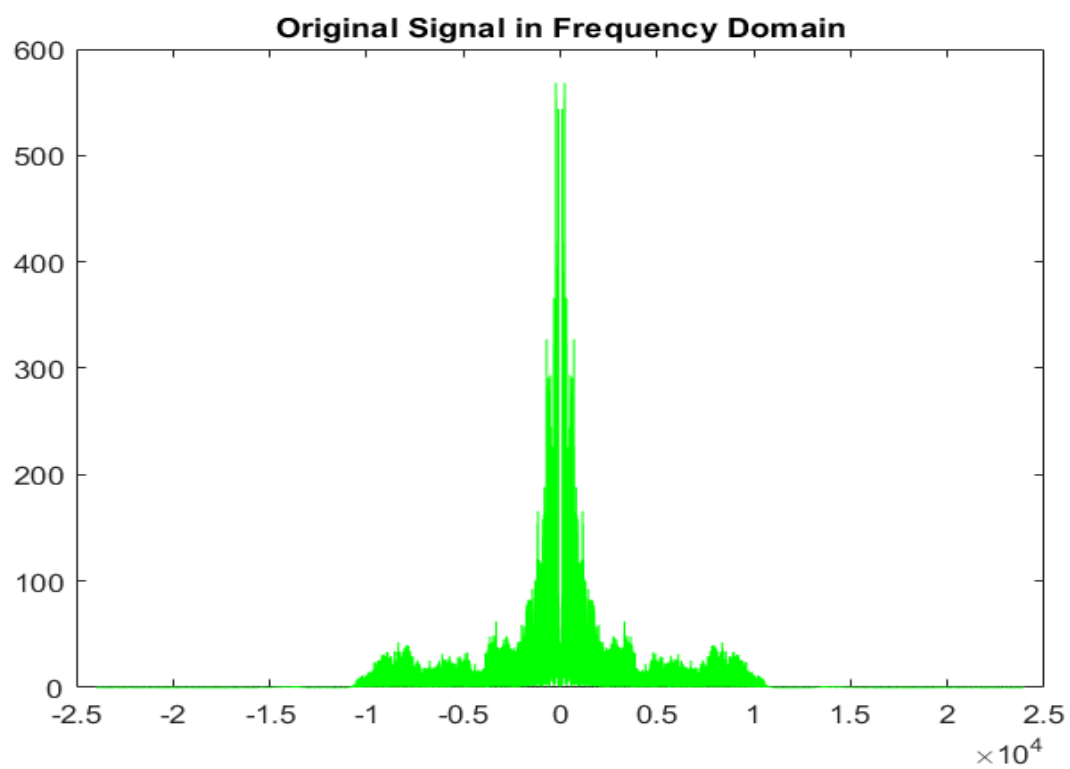
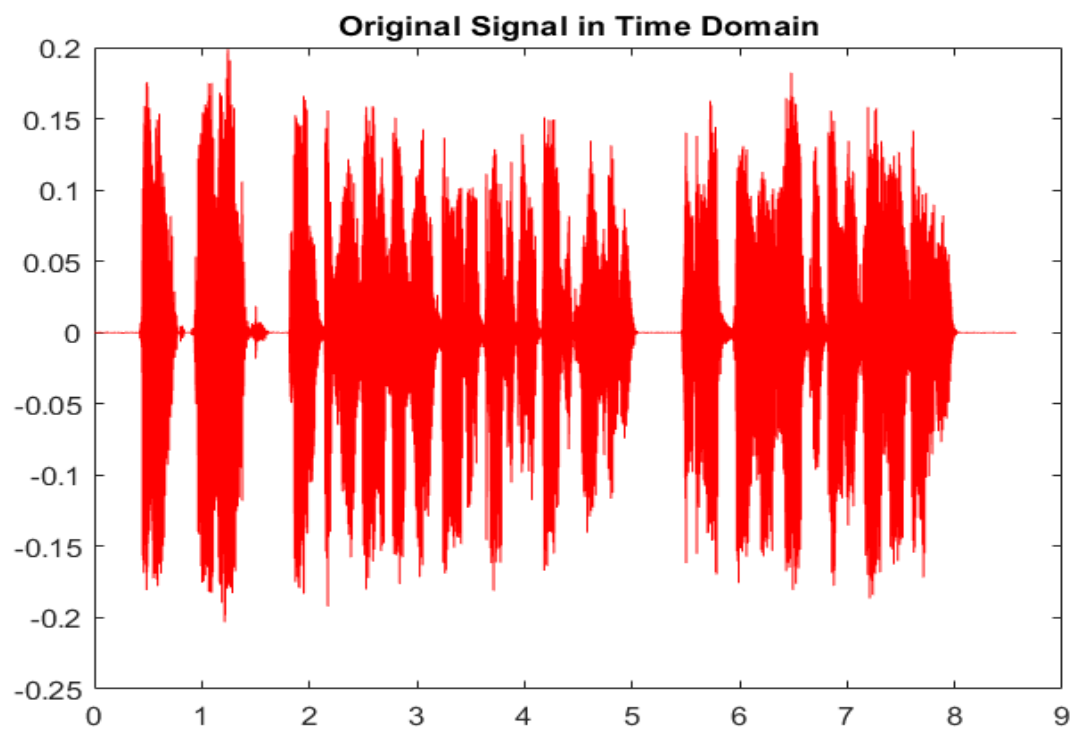
```

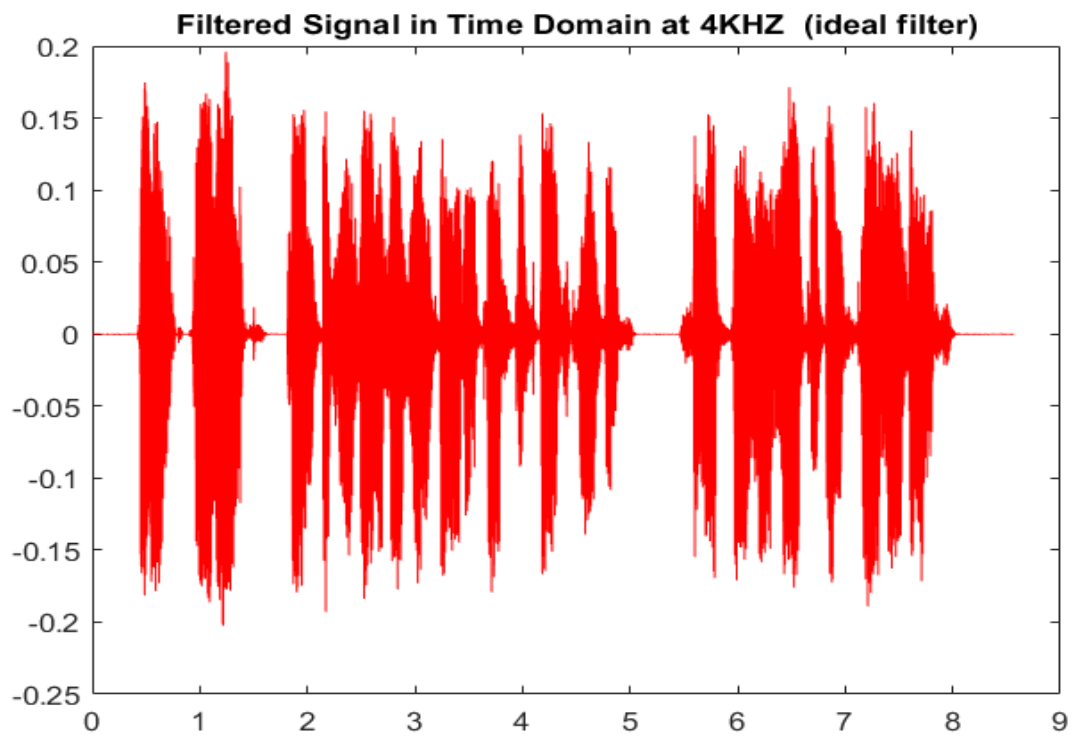
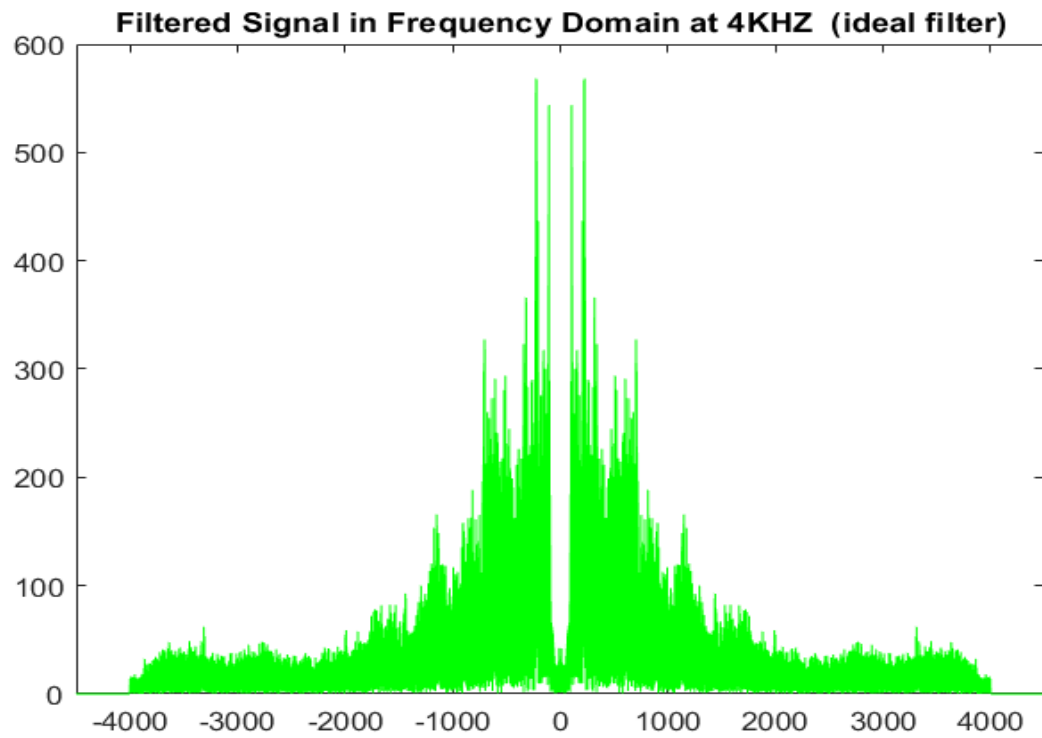
OBTAIN SSB-TC using ideal filter

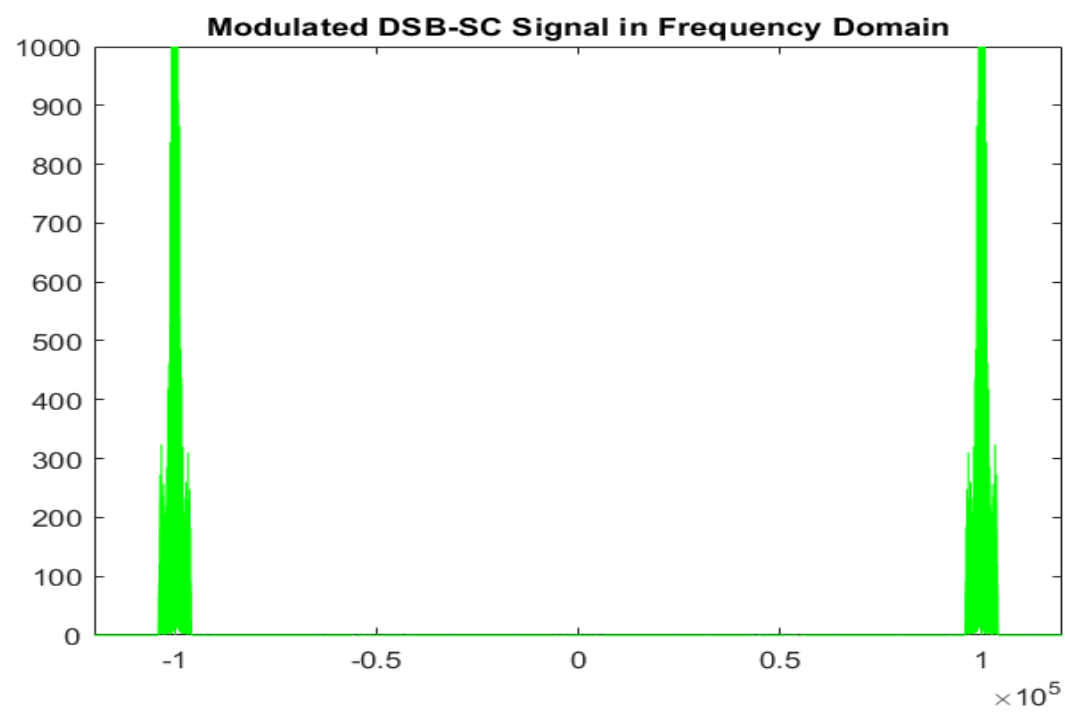
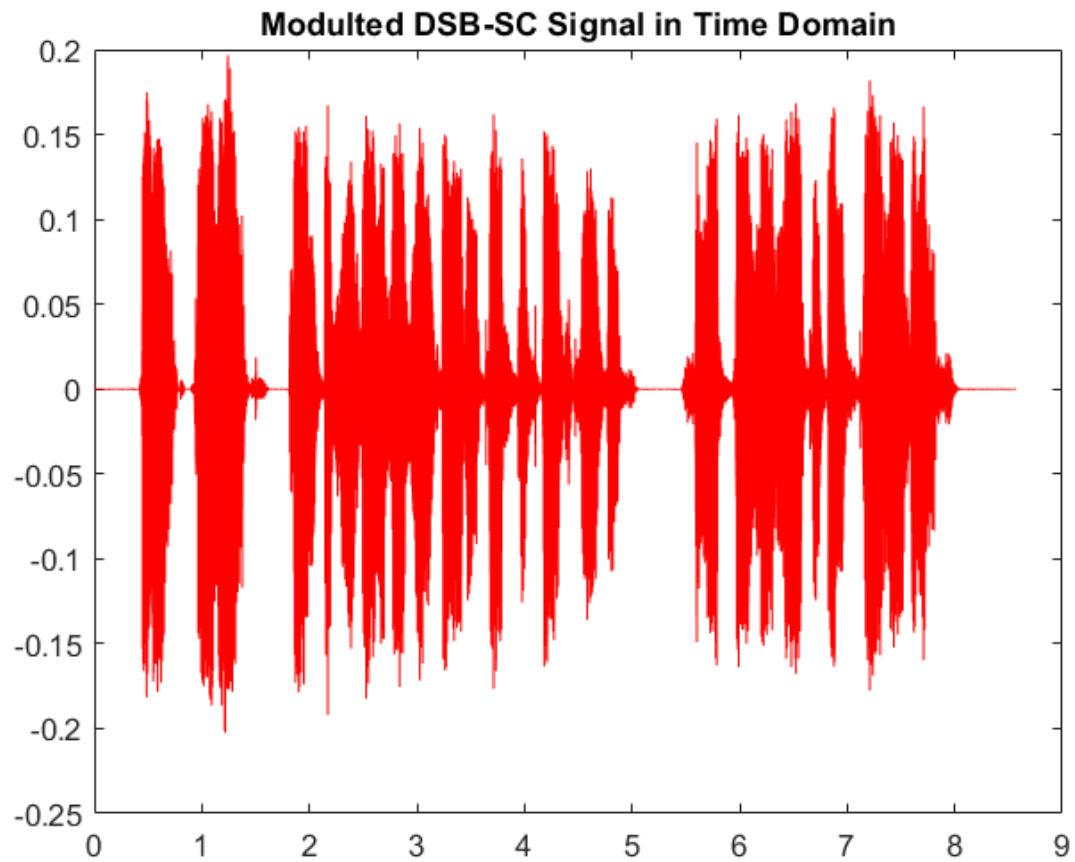
```
A=2*max(LSB_time);
carrier=carrier.';
SSB_TC=(A+LSB_time).*carrier;
SSB_TC_freq = fftshift(fft(SSB_TC));
step=step.';
step=resample(step,500000,F);
SSB_SCinFreqIdeal = SSB_TC_freq.*step;
SSB_TC_time = ifft(ifftshift(SSB_SCinFreqIdeal));
envelopeSSBTC=abs(hilbert(real(SSB_TC_time)));

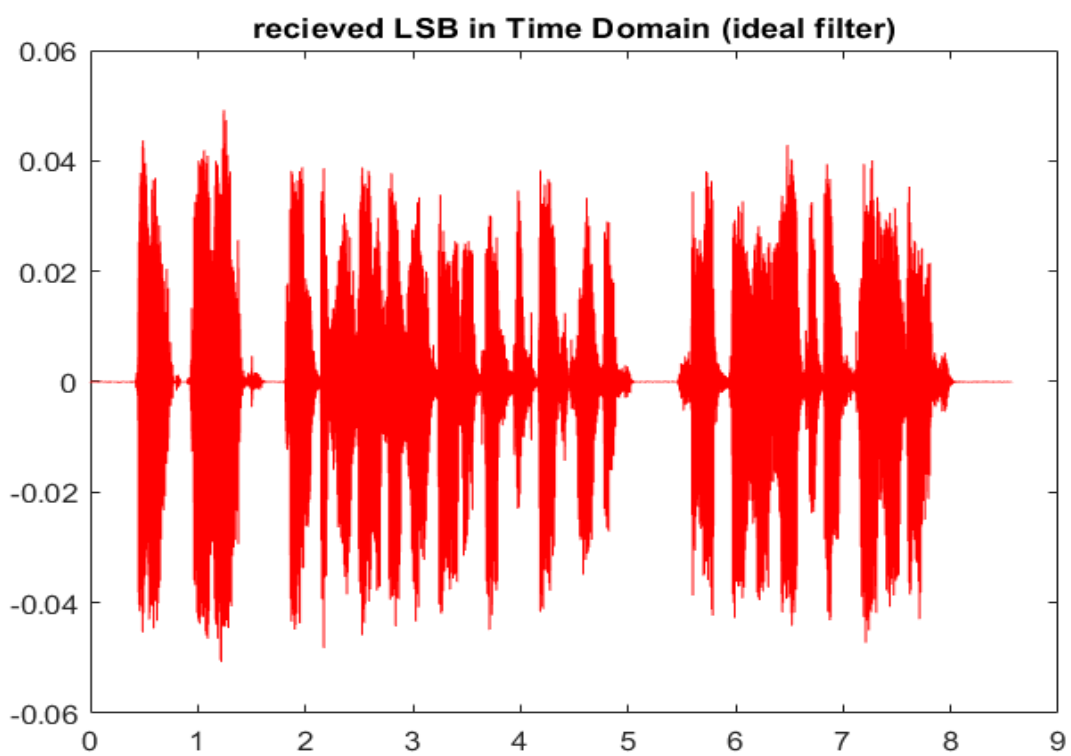
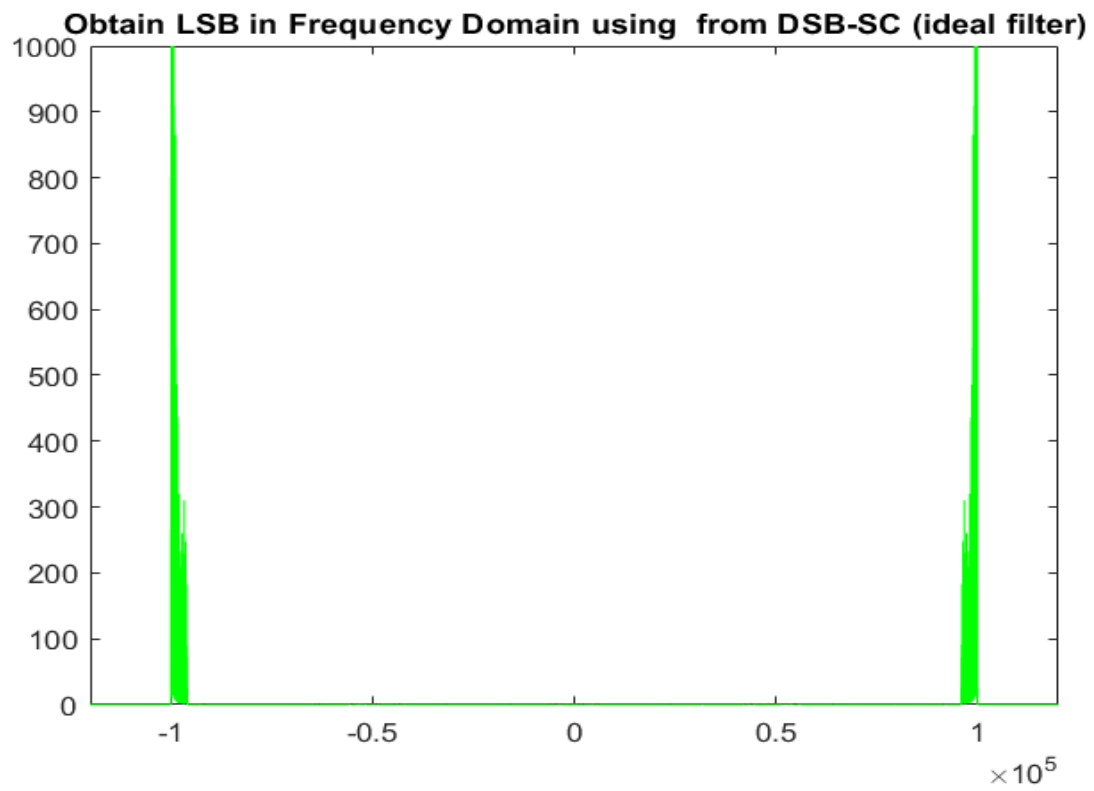
%down sample envelop detector
envelopeSSBTC=resample(envelopeSSBTC,F,500000);
envelopeSSBTC=envelopeSSBTC-mean(envelopeSSBTC);
figure;
plot(T,envelopeSSBTC(1:end-1),'r');
title('Demodulated SSB-TC Signal in Time Domain');
zdtc = abs(fftshift(fft(envelopeSSBTC)));
fdtc = linspace(-F/2, F/2, length(envelopeSSBTC));
figure
plot(fdtc, zdtc,'g');
xlim([-5000 5000]);
ylim([0 300]);
title('Demodulated SSB-TC Signal in Frequency Domain');
sound(real(envelopeSSBTC),F);
```

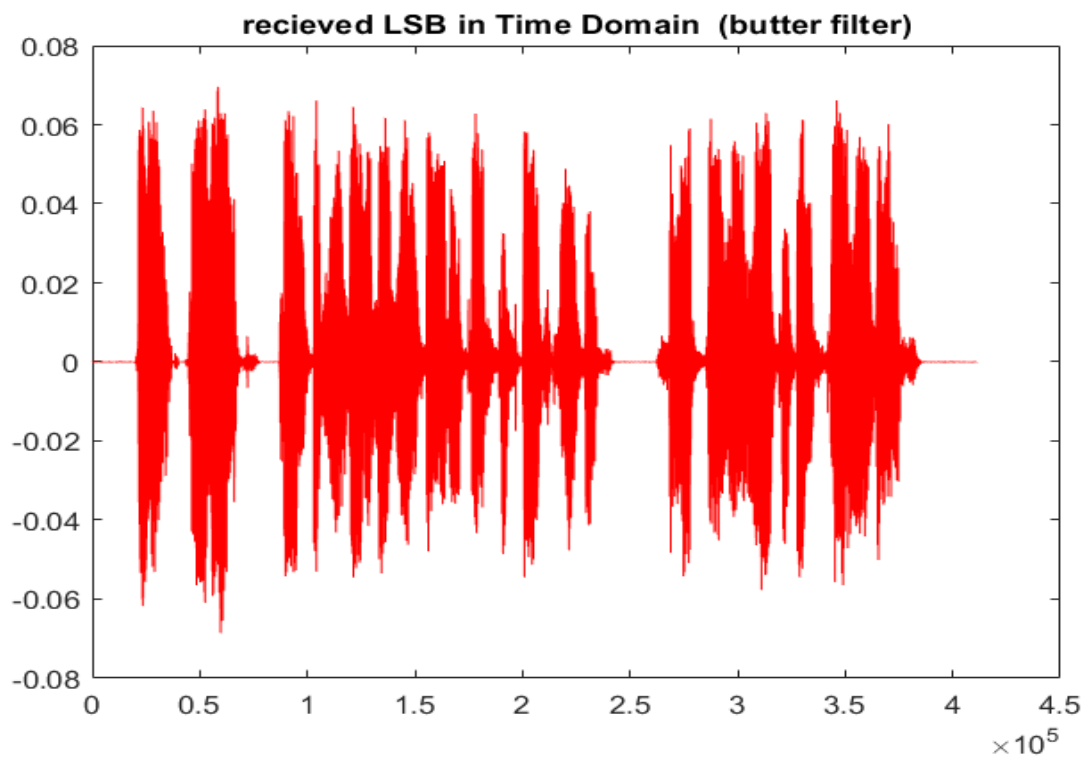
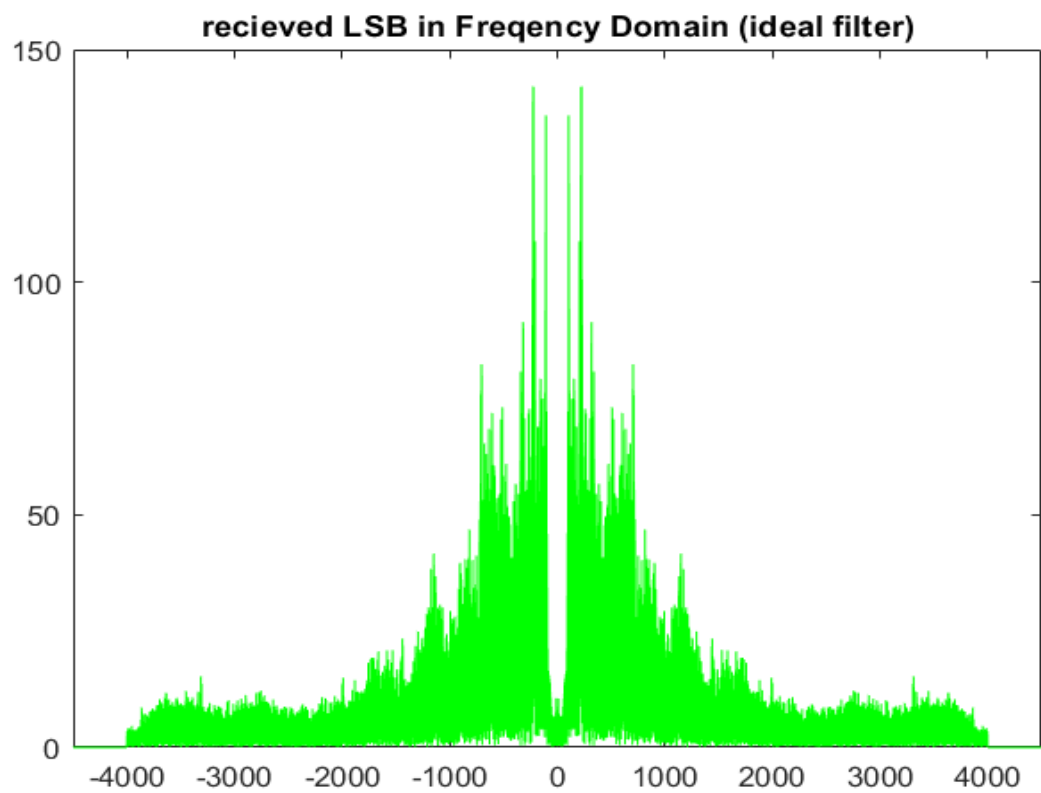

output:

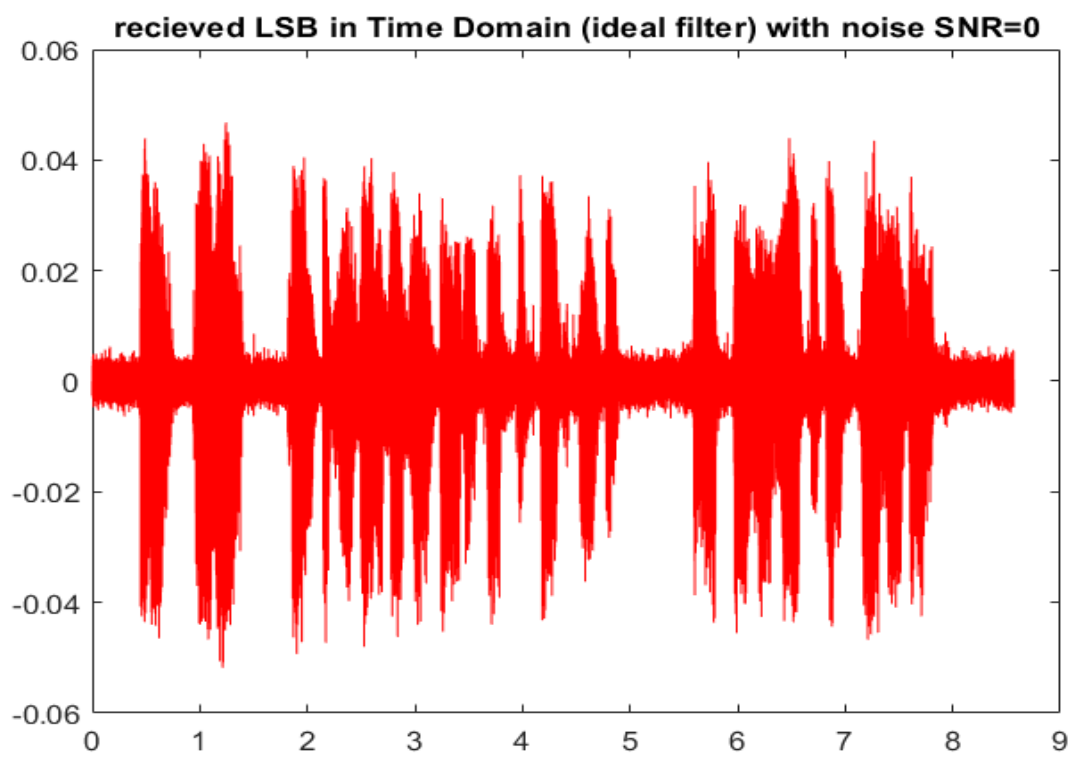
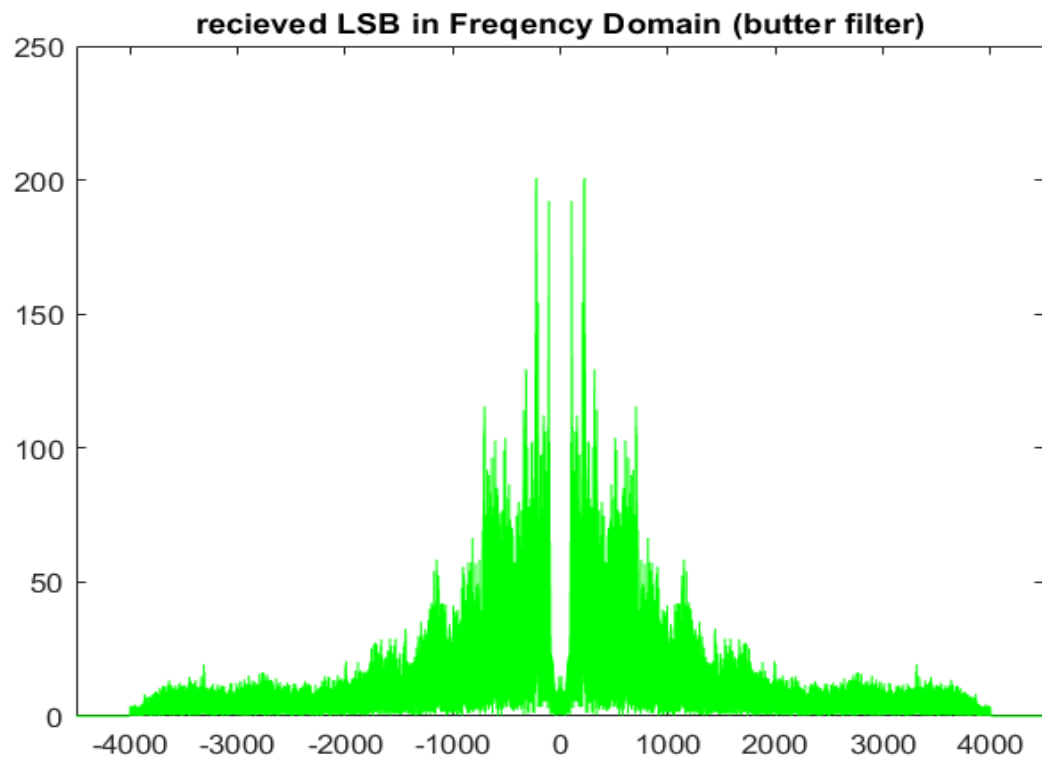


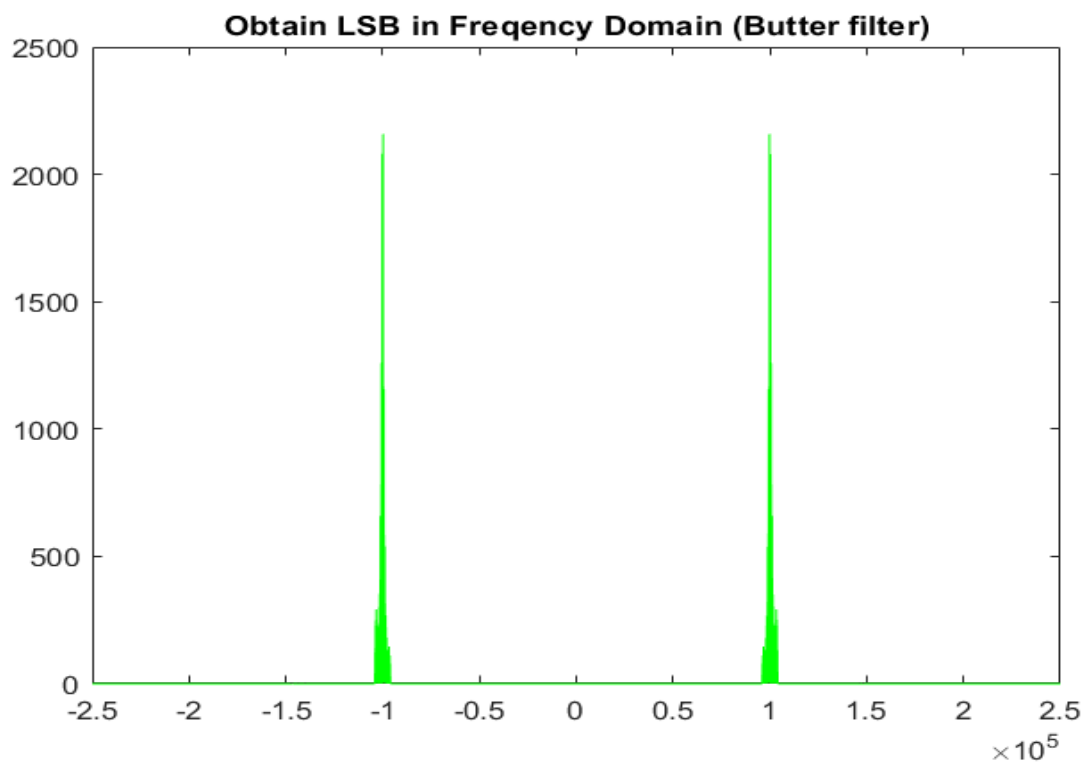
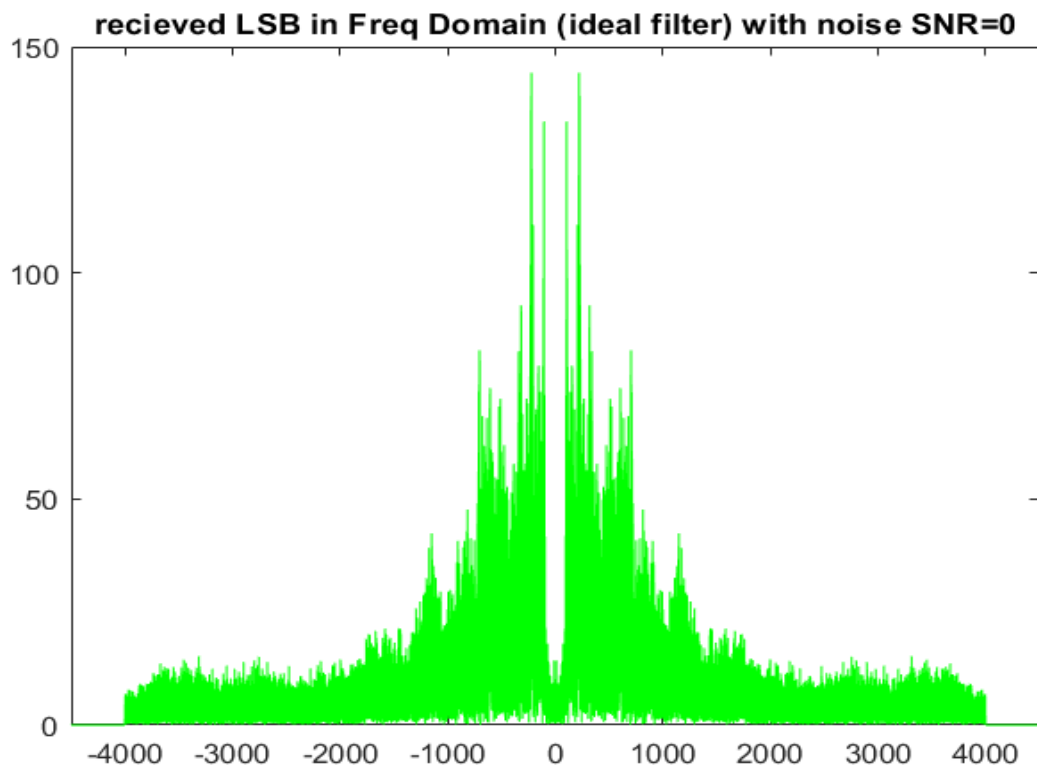


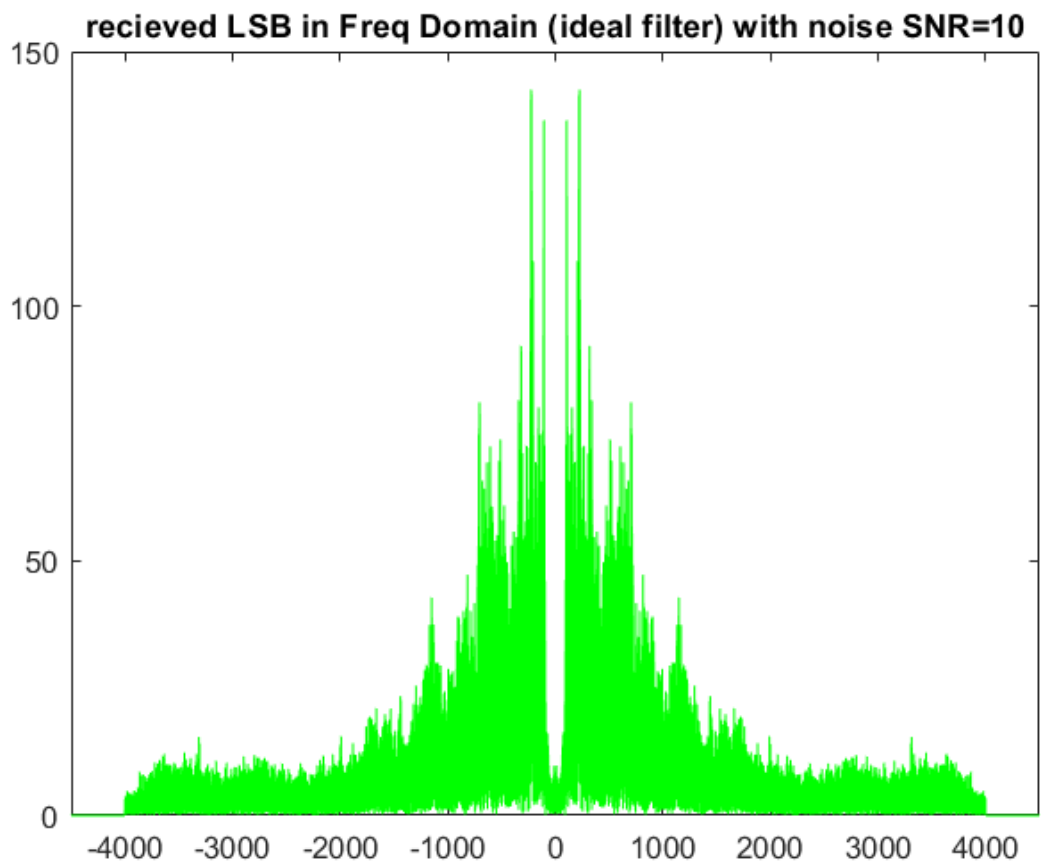
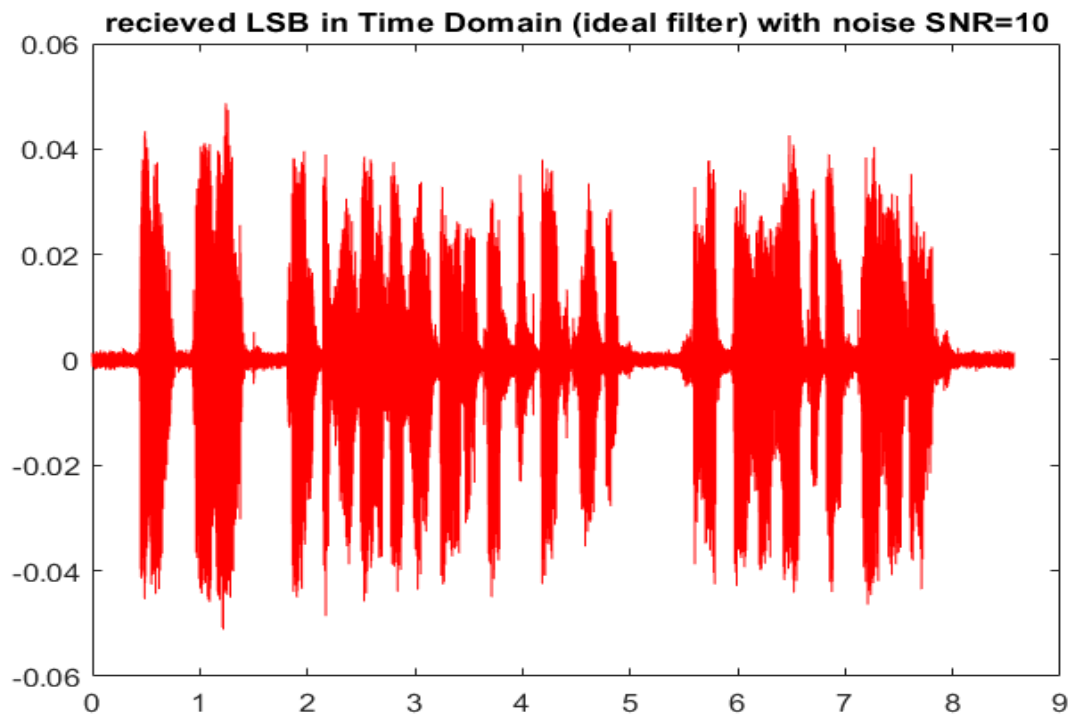


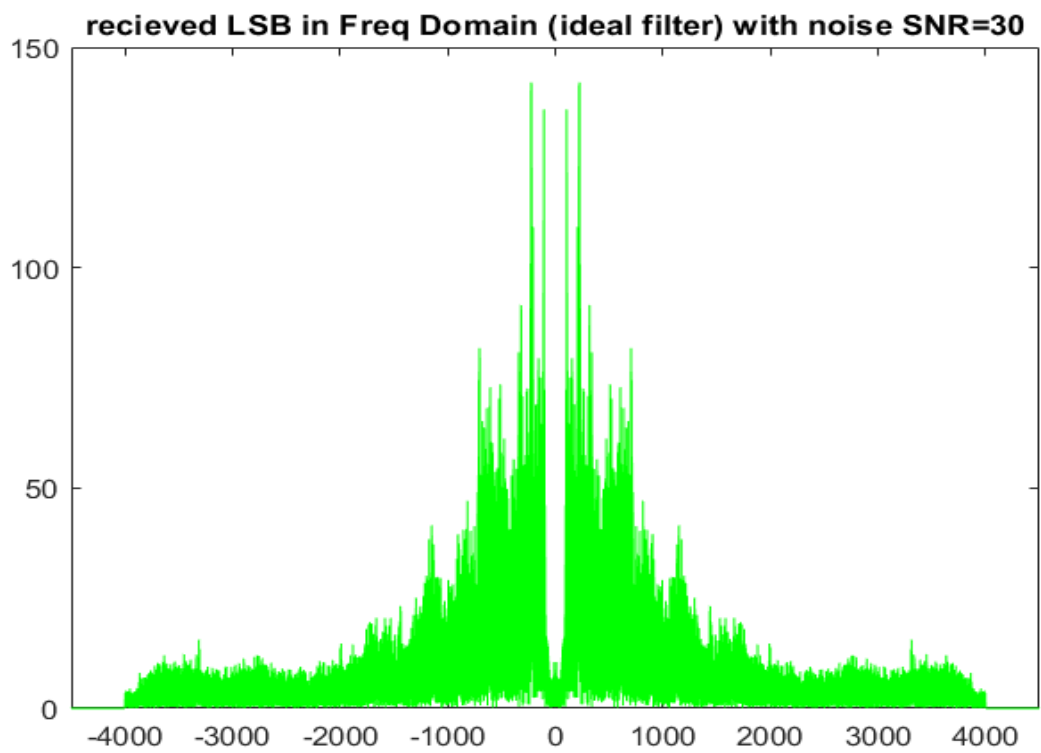
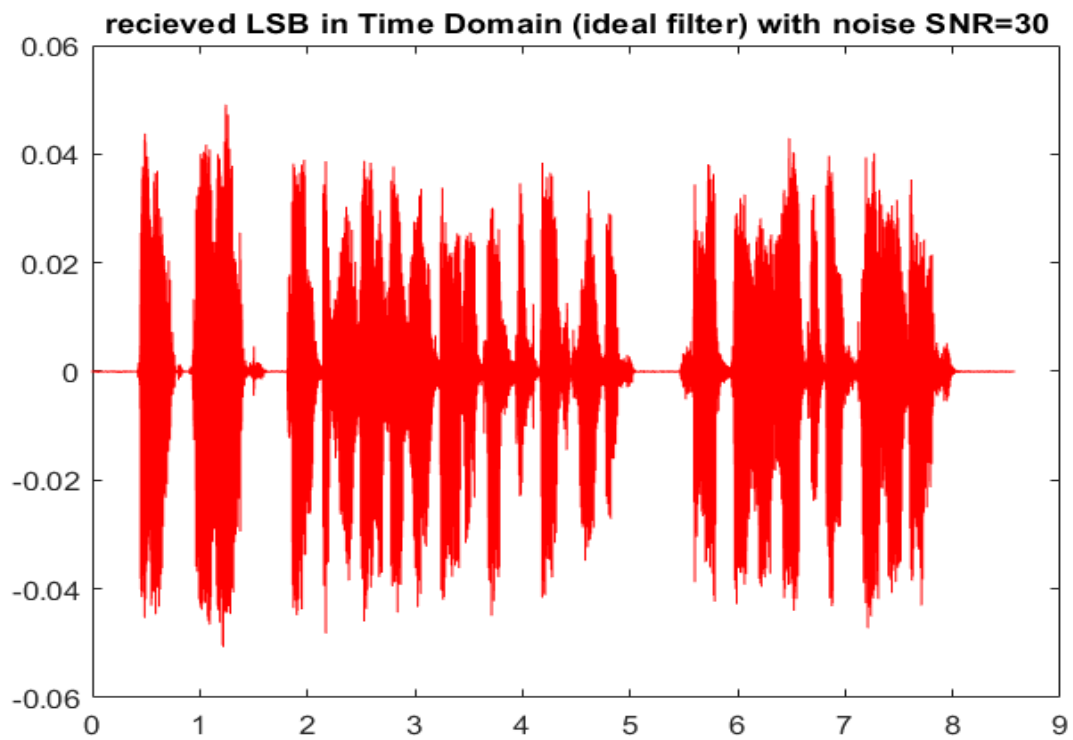


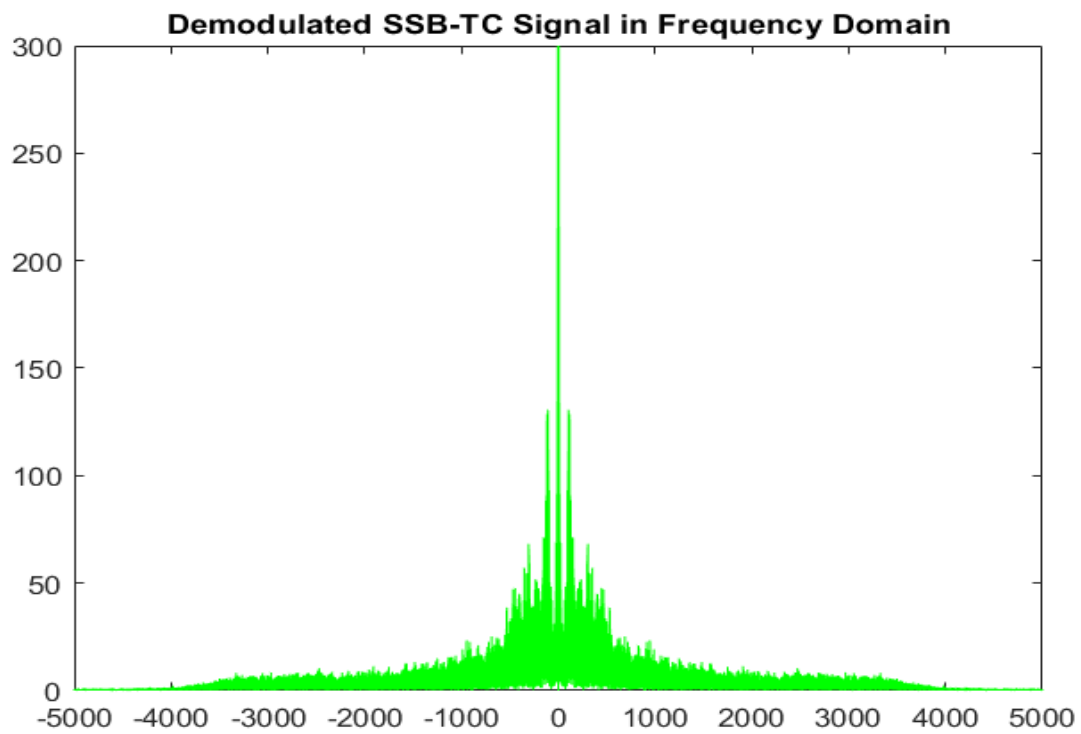
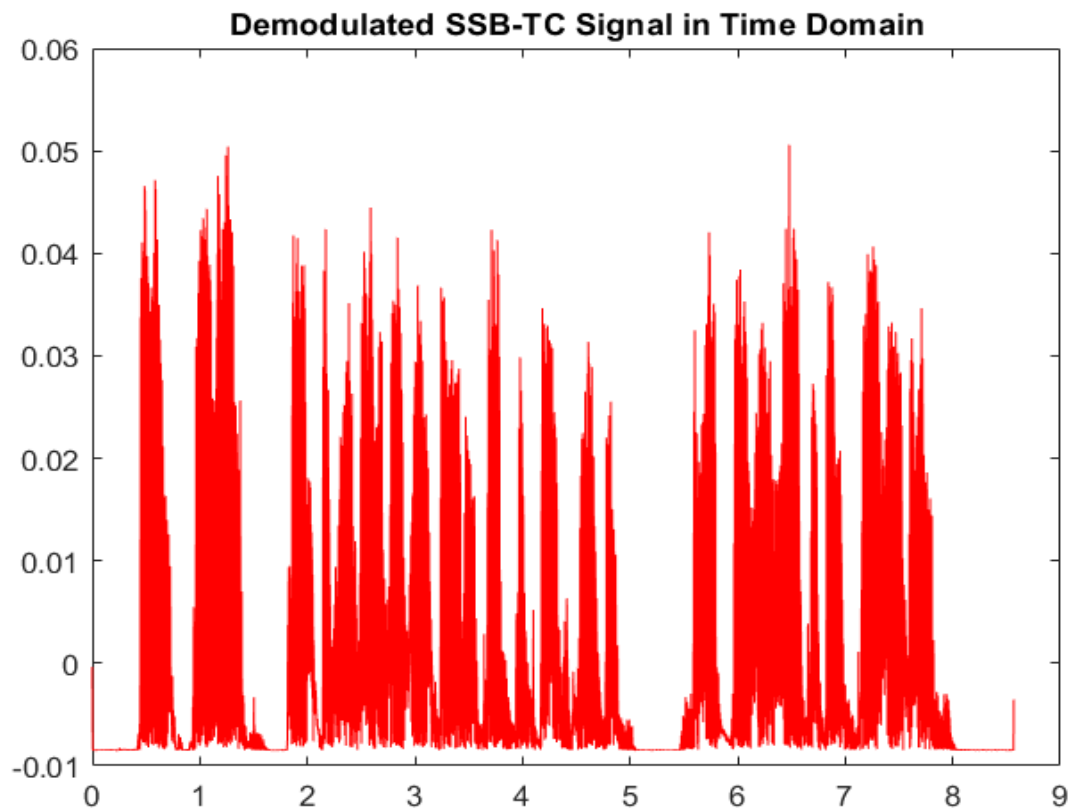












Experiment 3:

Code:

```
clc
clear
close all
```

Read the audio file and get its sampling frequency

```
[y, F] = audioread('eric.wav');

ty = length(y)/F;
t = linspace(0, ty, length(y));
```

Input signal conversion to frequency domain (to apply ideal LPF)

```
z = fftshift(fft(y));
f = linspace(-F/2, F/2, length(y));

figure
plot(f, abs(z), 'g');
title('Original Signal in Frequency Domain');
```

Searching for indices where $f = -4000$ and $f = 4000$

```
for i = 1:length(f)
    if (abs((f(i)+4.0000e+3)) < 0.01)
        index1 = i;
    end
    if (abs((f(i)-4.0000e+3)) < 0.01)
        index2 = i;
        break;
    end
end

% Generating a rect from index1 to index2
range = index2-index1;
step = [zeros(1, index1) ones(1, range) zeros(1, length(f)-index2)];
step = step.';
```

Multiplying input signal by rect to eliminate frequencies other than 4k

```
yfilteredFreq = step.*z;  
yfilteredFreqAbs = abs(step.*z);  
  
figure  
plot(f, yfilteredFreqAbs, 'g');  
xlim([-15000 15000]);  
title('Filtered Signal in Frequency Domain');
```

Converting the filtered signal to time domain to be modulated

```
yfiltered = ifft(fftshift(yfilteredFreq));  
figure  
plot(t, yfiltered, 'r');  
title('Filtered Signal in Time Domain');
```

Signal after low pass filter

```
sound(yfiltered, F);  
pause(9);
```

Filtered signal resampling

```
maxAmplitude = max(yfiltered);  
yfiltered = resample(yfiltered, 500000, F);  
yfiltered = yfiltered.';  
interval = length(yfiltered);
```

Frequency Modulation

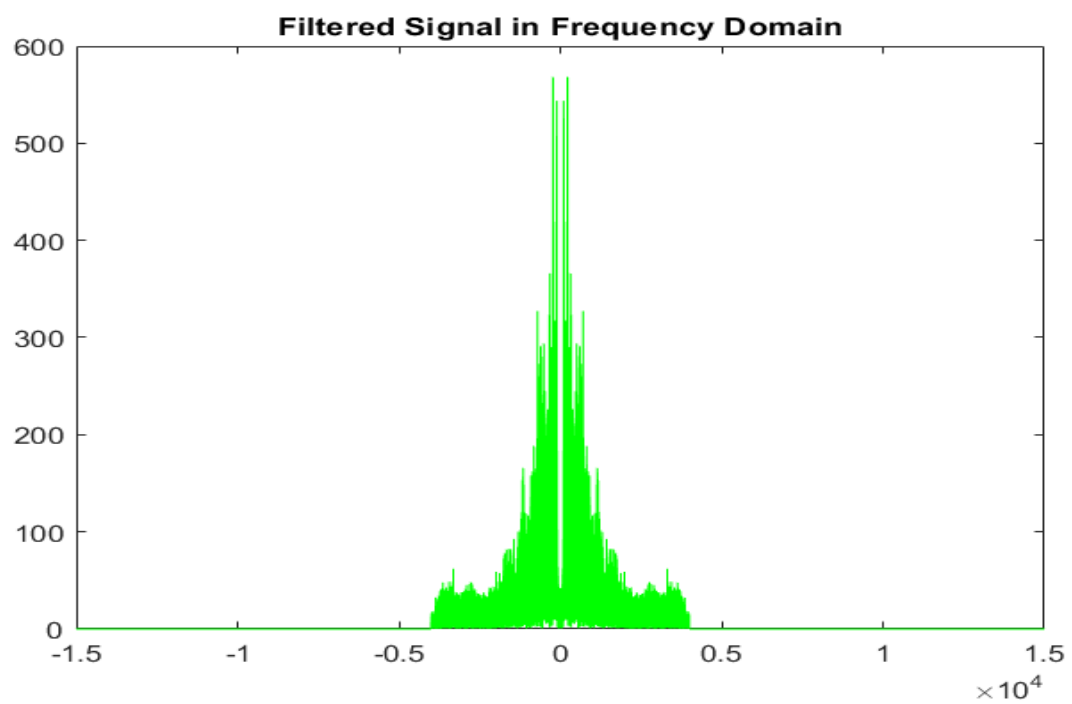
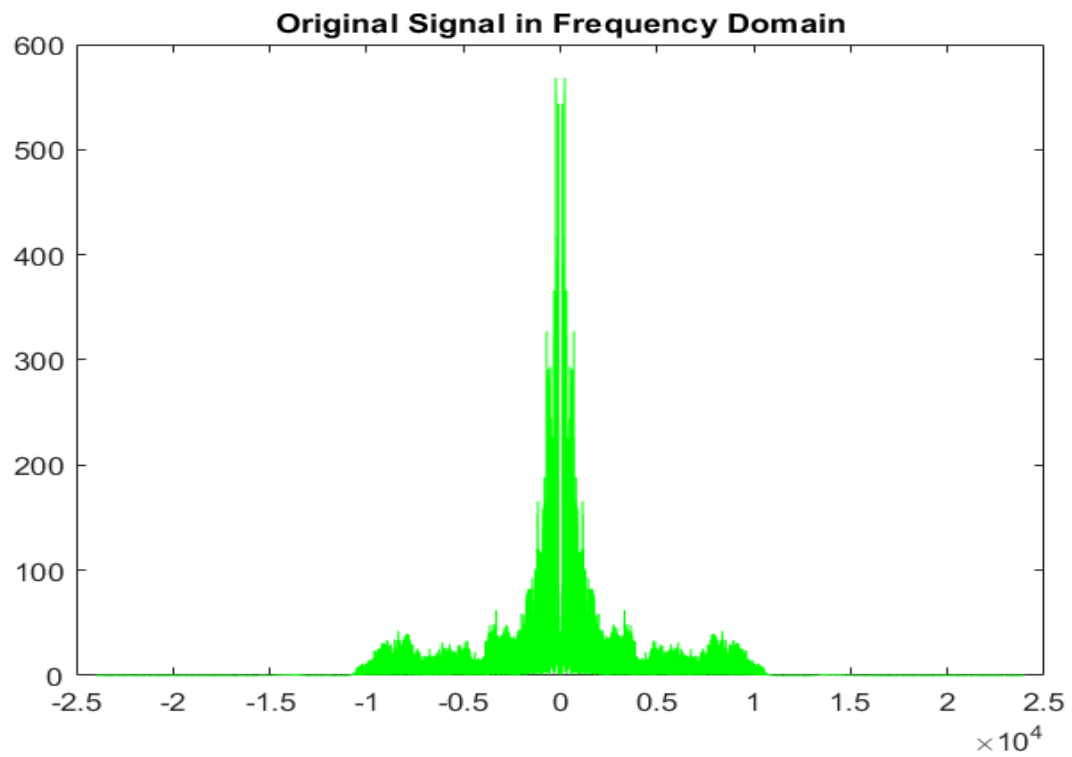
NBFM Signal

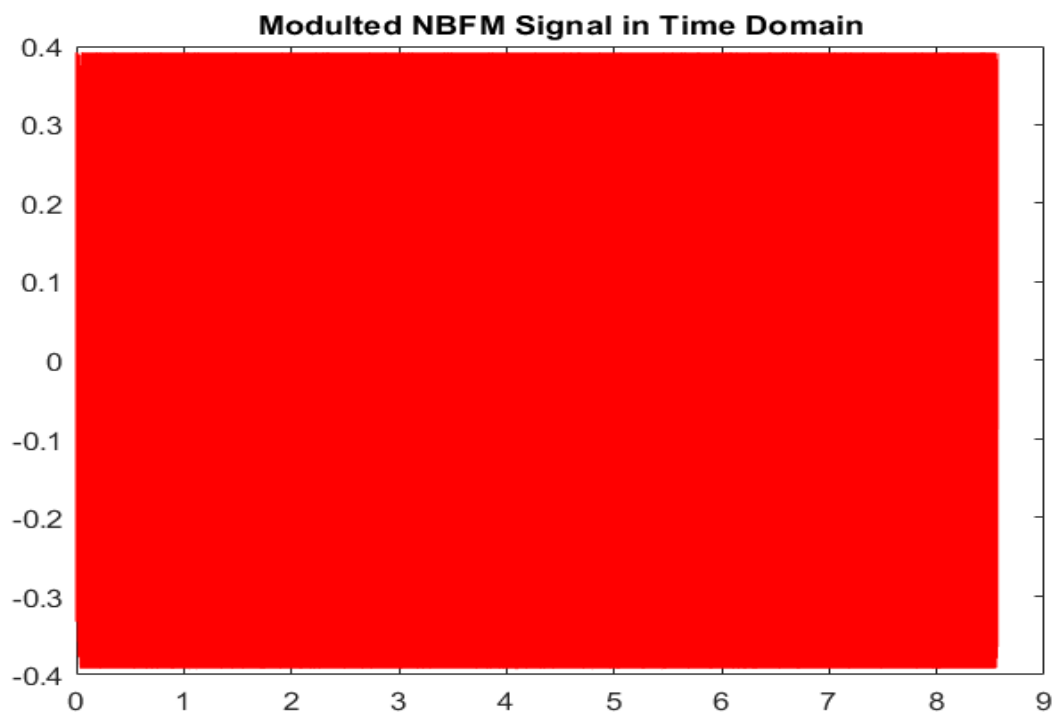
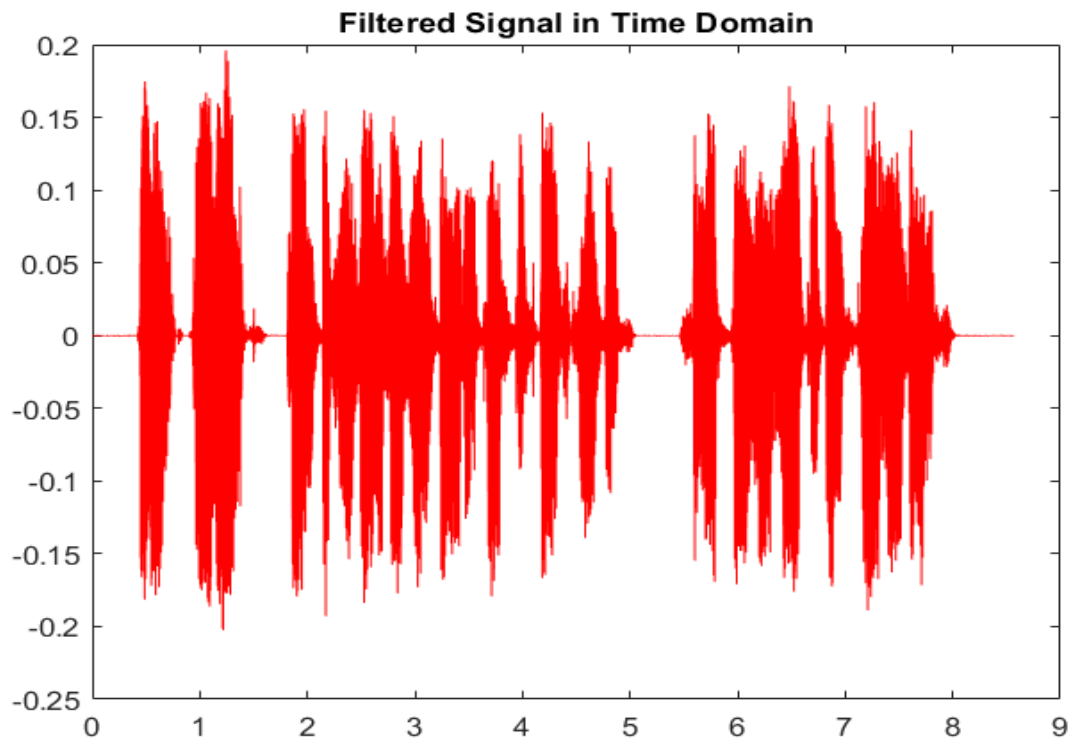
```
kf = 1e+1*pi;  
t = linspace(0, ty, interval);  
deltaF = abs(kf*max(yfiltered));  
m_int = kf.*cumsum(yfiltered);  
nbfm = 2*maxAmplitude.*cos(2.*pi.*100000.*t + m_int);  
figure  
plot(t, nbfm, 'r');  
title('Modulated NBFM Signal in Time Domain');  
zmnbfm = abs(fftshift(fft(nbfm)));  
fmnbfm = linspace(-500000/2, 500000/2, length(nbfm));  
figure  
plot(fmnbfm, zmnbfm, 'g');  
title('Modulated NBFM Signal in Frequency Domain');
```

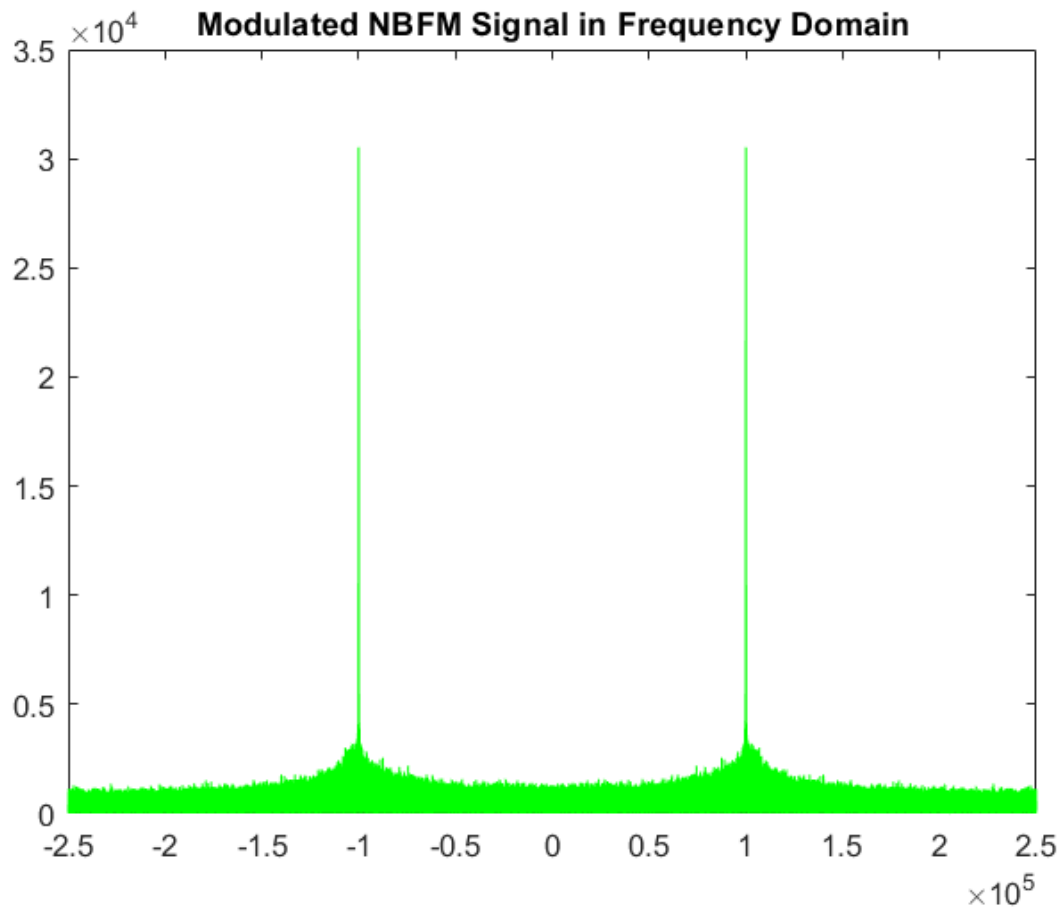
Frequency Demodulation

```
nbfm = diff(nbfm);  
nbfmEnvelope = abs(hilbert(nbfm));  
nbfmEnvelope = resample(nbfmEnvelope, F, 500000);  
sound(nbfmEnvelope , F);
```

output:







Experiment 3 Conclusions

- When the instantaneous frequency increases the value of the phase deviation increases.
- To achieve NBFM modulation index, frequency deviation, phase deviation and frequency deviation constant are very small.