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#### Read in the file

```
[m, Fs] = audioread('test_message.mp3');
T = 1 / Fs; L =
length(m); t =
(0:L-1) * T;
```

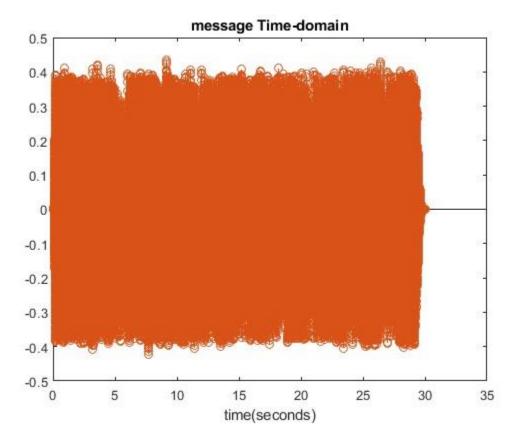
### Create object for playing audio

```
pl = audioplayer(m,Fs); % original signal
%pl.play;
```

### Plot audio

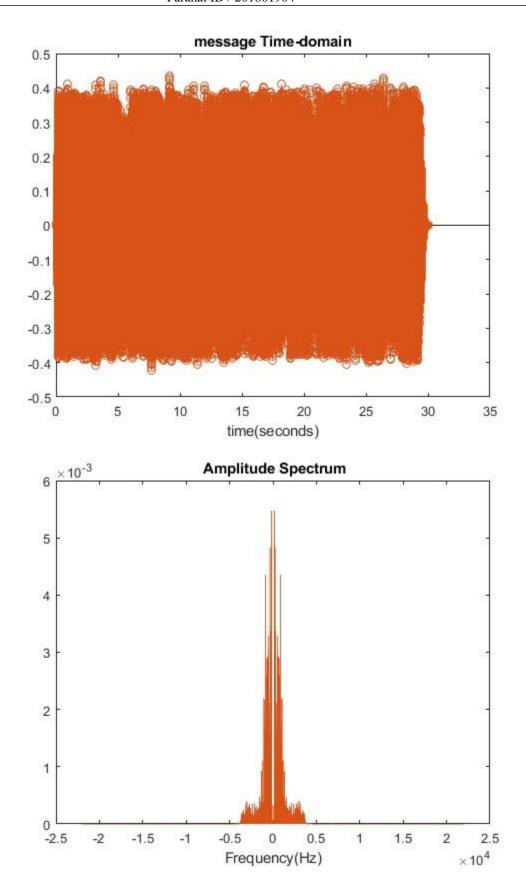
```
N = size(m, 1); figure;
stem(t, m);
```





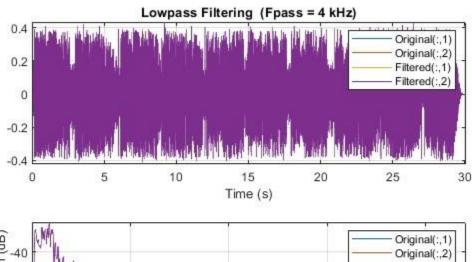
### Plot the spectrum

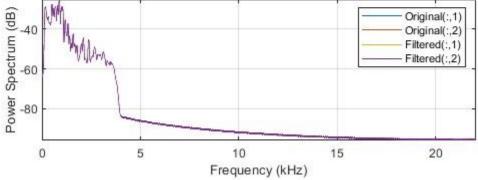
```
dm = Fs / N; w = (-(N/2):(N/2)-1)*dm; y = fft(m) / N;
% For normalizing y2 = fftshift(y); figure; plot(w,
abs(y2)); title(' Amplitude Spectrum');
xlabel('Frequency(Hz)');
```



### filtering at 4K Hz (Low pass filter)

lowpass(m,4000,Fs); filter\_m =
lowpass(m,4000,Fs);





# the average error between the original signal and the band limited signal.

```
err = immse(m,filter_m);
disp("the error =") disp
(err)

the error =
5.2813e-09
```

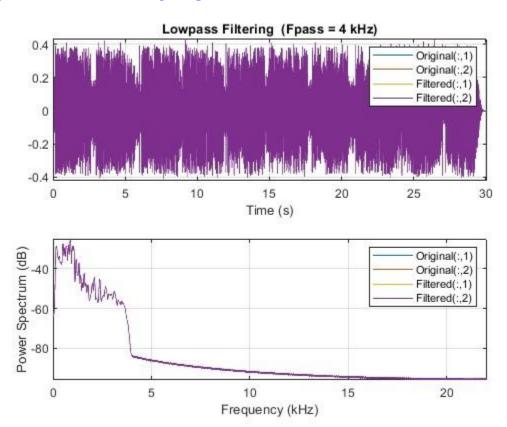
### AM modulation of the recorded signal

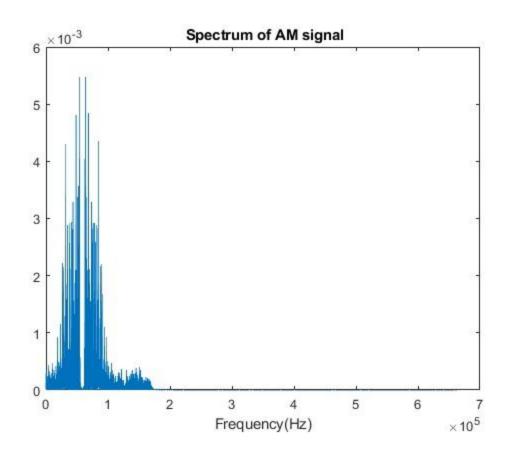
AM modulate signal = ammod(m,Fs,1000000);

### Spectrum of modulated signal

spectrumAM = fft(AM modulate signal);

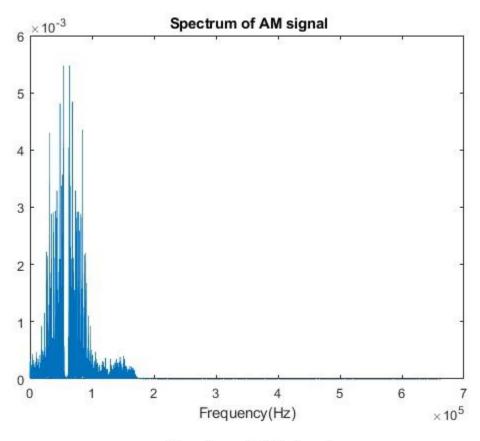
lengthOfSignal = length(m); normalizedSpectrumAm=
abs(spectrumAM/lengthOfSignal); frequencySpectrumAm=
normalizedSpectrumAm(1:lengthOfSignal/2+1);
frequencySpectrumAm(2:end-1) = 2\*frequencySpectrumAm(2:end-1);
figure; plot(frequencySpectrumAm); title(' Spectrum of AM signal'); xlabel('Frequency(Hz)');

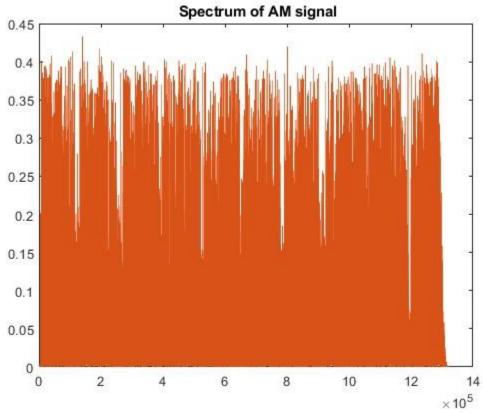




# **Envelope Detection based on Hilbert Transform and then FFT**

```
envelope = abs (hilbert (AM_modulate_signal));
figure; plot(envelope); title(' Spectrum of AM signal');
```





# the average error between the original signal and the transmitted message.

```
err = immse(m,envelope);
disp("the error =") disp
(err)

the error =
0.0423
```

### signal with noise

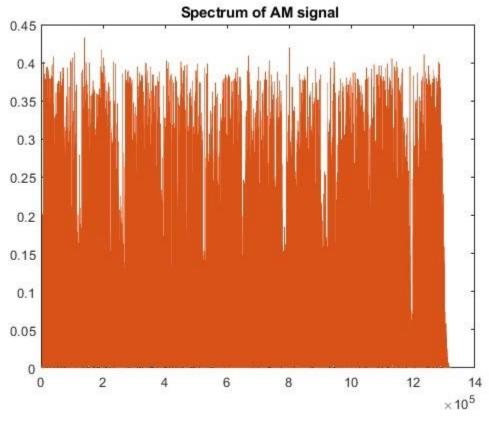
```
Noise signal = awgn(m, 10);
```

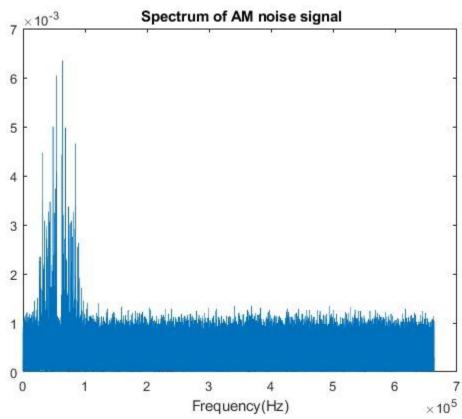
### AM modulation of the recorded signal + Noise

```
AM modulate noise signal = ammod(Noise signal, Fs, 1000000);
```

### Spectrum of modulated signal

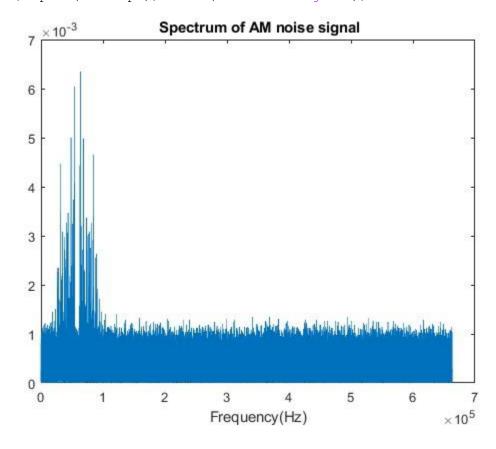
```
spectrumAM = fft(AM_modulate_noise_signal); lengthOfSignal =
length(m); normalizedSpectrumAm= abs(spectrumAM/lengthOfSignal);
frequencySpectrumAm= normalizedSpectrumAm(1:lengthOfSignal/2+1);
frequencySpectrumAm(2:end-1) = 2*frequencySpectrumAm(2:end-1);
figure; plot(frequencySpectrumAm); title(' Spectrum of AM noise
signal'); xlabel('Frequency(Hz)');
```

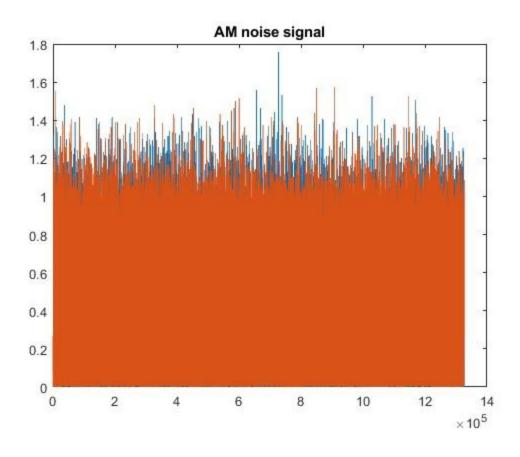




# **Envelope Detection based on Hilbert Transform and then FFT for noise signal**

envelope = abs (hilbert (AM\_modulate\_noise\_signal));
figure; plot(envelope); title('AM noise signal');





## the average error between the original signal and the transmitted message.

```
err = immse(m,envelope);
disp("the error =") disp
(err)

the error =
0.1421
```

## FM modulatiom Read in the file

```
[m, Fs] = audioread('test_message.mp3');
T = 1 / Fs; L =
length(m); t =
(0:L-1) * T;
```

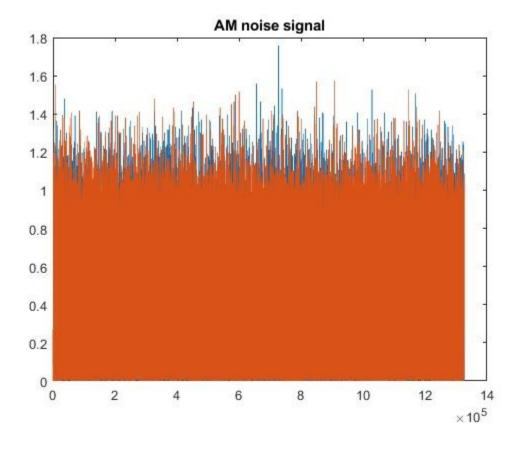
### Create object for playing audio

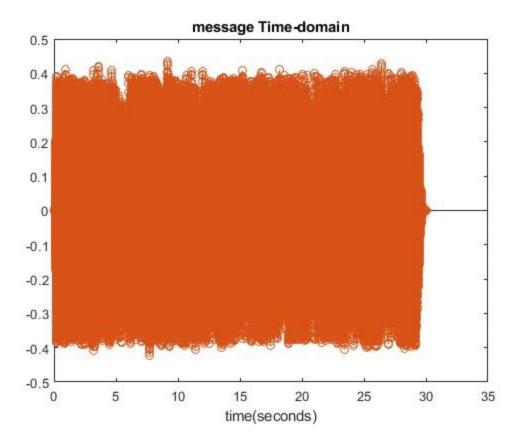
```
pl = audioplayer(m,Fs); % original signal
```

%pl.play;

### Plot audio

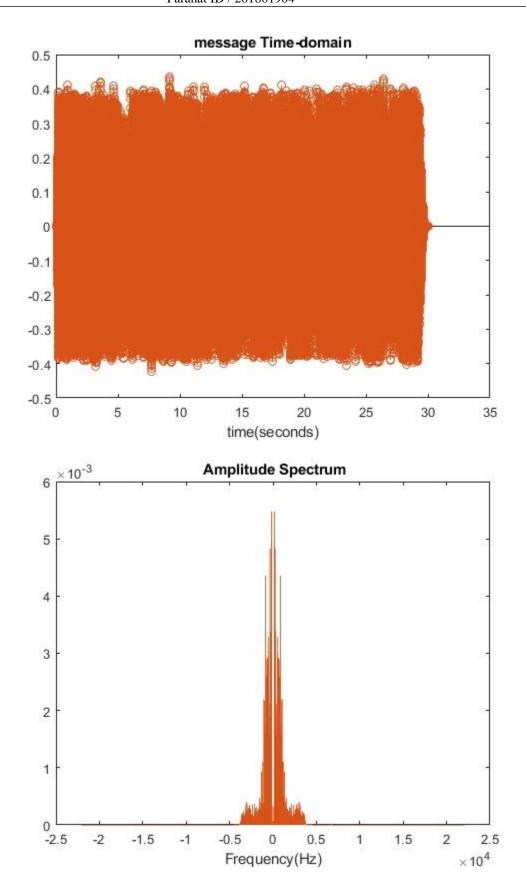
N = size(m, 1); figure; stem(t, m); title('message Timedomain'); xlabel('time(seconds)');





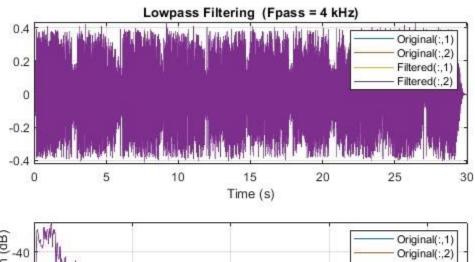
### Plot the spectrum

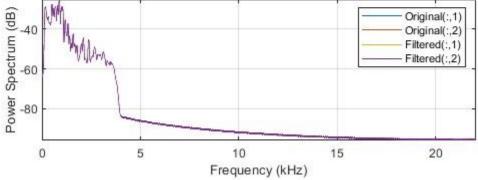
```
dm = Fs / N; w = (-(N/2):(N/2)-1)*dm; y = fft(m) / N;
% For normalizing y2 = fftshift(y); figure; plot(w,
abs(y2)); title(' Amplitude Spectrum');
xlabel('Frequency(Hz)');
```



### filtering at 4K Hz (Low pass filter)

lowpass(m,4000,Fs); filter\_m =
lowpass(m,4000,Fs);





# the average error between the original signal and the band limited signal.

```
err = immse(m,filter_m);
disp("the error =") disp
(err)

the error =
5.2813e-09
```

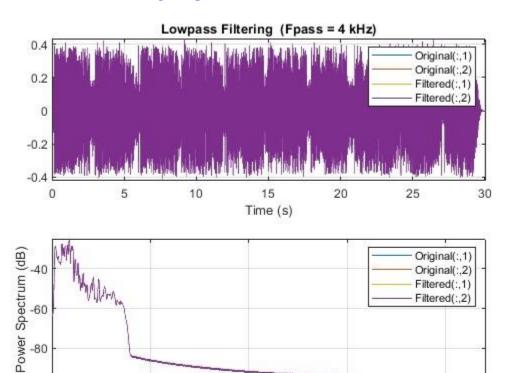
### FM modulation of the recorded signal

FM modulate signal = fmmod(m,Fs,1000000,2);

### Spectrum of modulated signal

spectrumFM = fft(FM modulate signal);

lengthOfSignal = length(m); normalizedSpectrumFm=
abs(spectrumFM/lengthOfSignal); frequencySpectrumFm=
normalizedSpectrumFm(1:lengthOfSignal/2+1);
frequencySpectrumFm(2:end-1) = 2\*frequencySpectrumFm(2:end-1);
figure; plot(frequencySpectrumFm); title(' Spectrum of FM signal'); xlabel('Frequency(Hz)');



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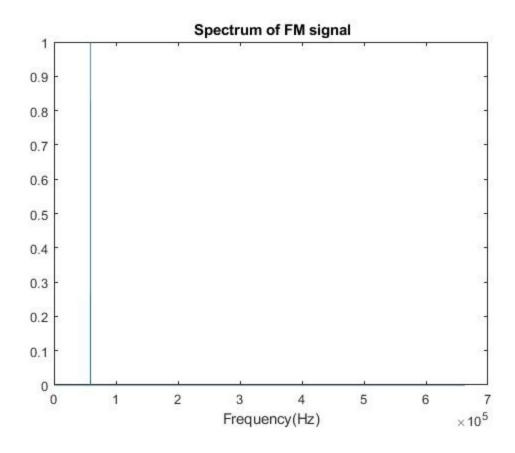
Frequency (kHz)

15

20

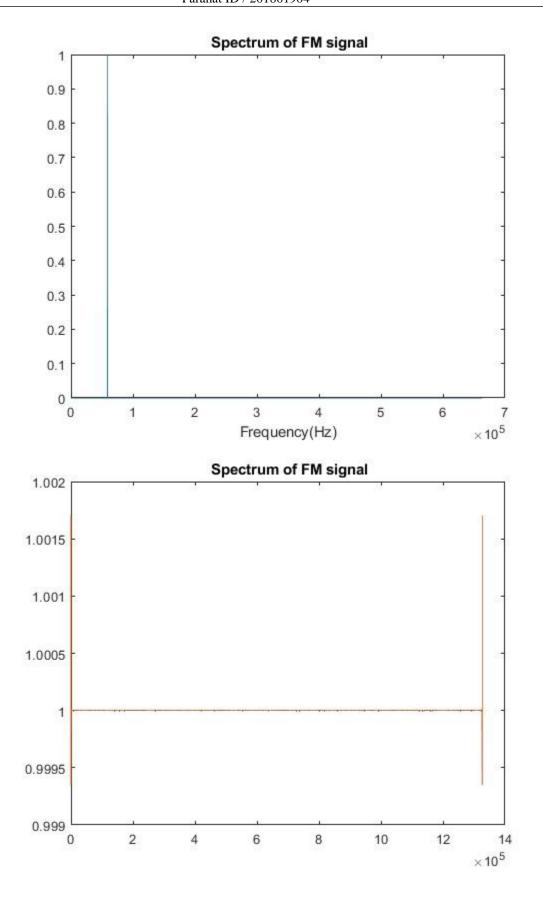
0

5



# **Envelope Detection based on Hilbert Transform and then FFT**

```
envelope = abs (hilbert (FM_modulate_signal));
figure; plot(envelope); title(' Spectrum of FM signal');
```



#### with noise

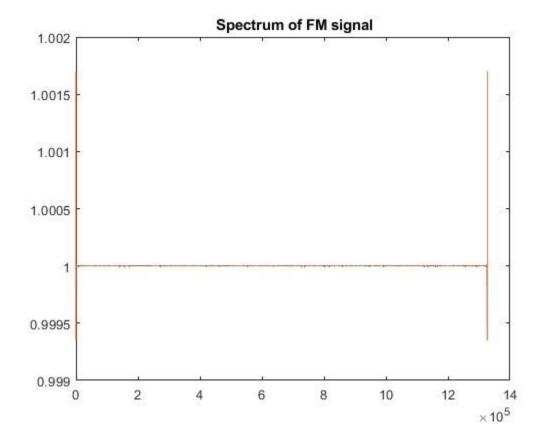
Noise signal = awgn(m, 10);

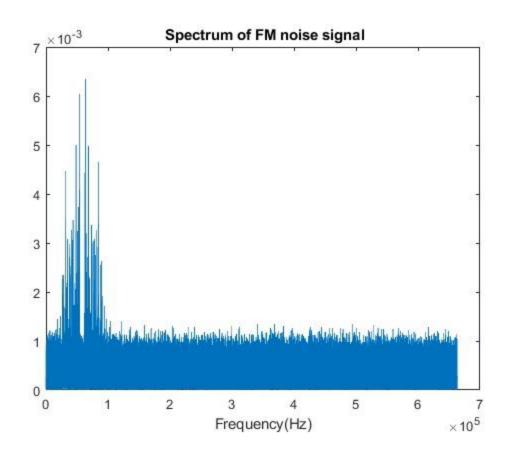
### AM modulation of the recorded signal + Noise

FM modulate noise signal = fmmod(Noise signal, Fs, 1000000, 2);

### Spectrum of modulated signal

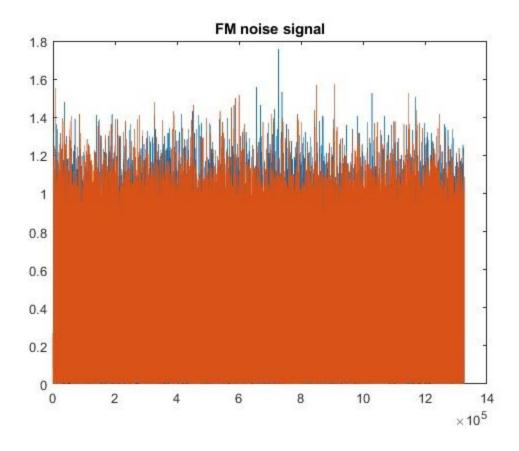
spectrumFM = fft(AM\_modulate\_noise\_signal); lengthOfSignal =
length(m); normalizedSpectrumFm= abs(spectrumFM/lengthOfSignal);
frequencySpectrumFm= normalizedSpectrumFm(1:lengthOfSignal/2+1);
frequencySpectrumFm(2:end-1) = 2\*frequencySpectrumFm(2:end-1);
figure; plot(frequencySpectrumFm); title(' Spectrum of FM noise
signal'); xlabel('Frequency(Hz)');





# **Envelope Detection based on Hilbert Transform and then FFT for noise signal**

```
envelope = abs (hilbert (AM_modulate_noise_signal));
figure; plot(envelope); title('FM noise signal');
```



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