CSE 3048 PROJECT: Demodulating A Composite Signal

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File: 81.txt

Modulation frequency 1: 5KHz Message 1: Trouble, Your Mind (not sure)

Modulation frequency 2: 10KHz Message 2: Guilty, Excellent

Explanation

Stage1: Taking composite signal and observing the FT of it.

First, I imported the file and took the Fourier transform of it. Then, I plot the results in both time and frequency domains.

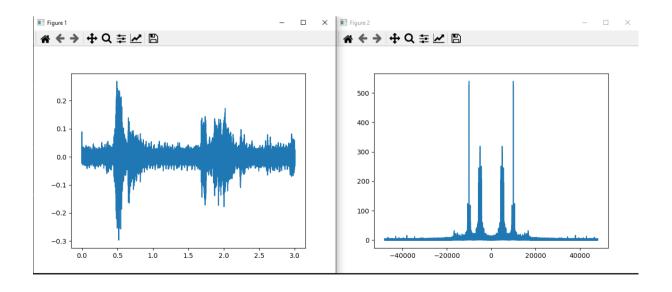
```
fs = 96000 # sample rate
seconds = 3 # recording duration
modulation_frequency = 5000

my_recording = np.loadtxt('81.txt')
time = np.linspace(0,seconds,my_recording.shape[0]) # start, stop, sample count

plt.figure(1) # composite signal in time domain
plt.plot(time, my_recording)

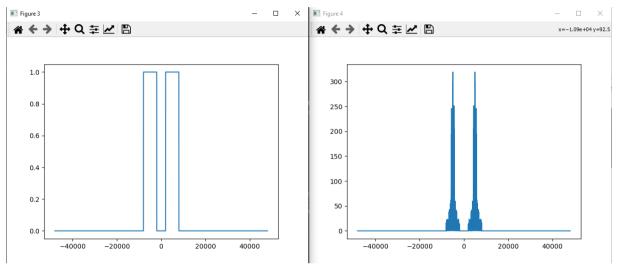
frequencies = np.linspace(-fs/2,fs/2,fs*seconds)
my_recording_in_frequency = fftshift(fft(my_recording))

plt.figure(2) # composite signal in frequency domain
plt.plot(frequencies, np.abs(my_recording_in_frequency))
```



Stage 2: Bandpass Filtering

When we observe the signal, we can see that their modulation frequencies are 5KHz and 10KHz. We should create a bandpass filter accordingly. We take +- 3 KHz of them. The creation of the bandpass filter and an example for the first signal (5KHz) are as follows.



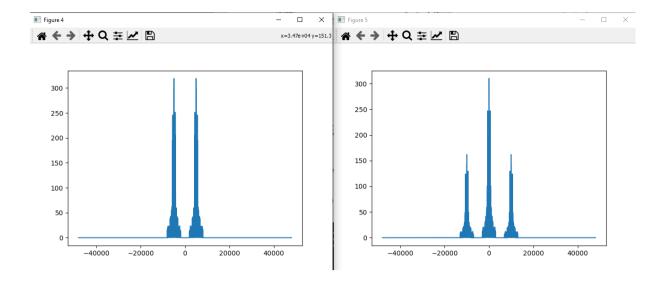
Stage 3: Multiplying with cos(2pift) in time domain.

I preferred multiplying my signal with cos(2pift) in time domain instead of convolving it in frequency domain. So, I took the inverse fourier transform of the signal and did the multiplication.

After that, I took its fourier transform again to progress. The code for these operations and plot results are as follows:

```
#stage2 : multiply with cos(2*pi*f*t) in time domain
bandpass_filtered_record_in_time = ifft(ifftshift(bandpass_filtered_record))
bandpass_filtered_record_stage2 = np.multiply(bandpass_filtered_record_in_time,np.cos(2*np.pi*modulation_frequency*time))
bandpass_filtered_record_stage2_in_frequency = fftshift(fft(bandpass_filtered_record_stage2))

plt.figure(5)
plt.plot(frequencies, np.abs(bandpass_filtered_record_stage2_in_frequency))
```



Stage 4: Lowpass Filtering

We only need the signal in the middle of the chart. So, I created a lowpass filter and used it.



Stage 5: Converting signal to a voice file.

After all the operations above, I took the inverse Fourier transform of the signal and tried to understand the message.

```
sf.write("5k.wav", np.abs(ifft(lowpass_filtered_record)), 96000, subtype= 'PCM_16')
```