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Problem 1

```
Fs = 8000;
Ts = 1/Fs;
t = 0:Ts:0.0005;    %0.5ms
f1 = 1209;
f2 = 1336;
y = cos(2*pi*f1*t) + cos(2*pi*f2*t);
```

1a Plot in time and freq domain

```
figure(1)
% Plot in time domain
plot(t, y), xlabel('Time (s)'), ylabel('Amplitude');
title('Signal y(t) in the Time Domain');
% Perform FFT and plot in the frequency domain
Y = fft(y);
frequencies = linspace(0, Fs, length(Y));
figure(2)
plot(frequencies, abs(Y)), xlabel('Frequency (Hz)'), ylabel('Magnitude');
title('Signal y(t) in the Frequency Domain');
xlim([0, Fs/2]);    %limit x-axis
```

1b Low Pass Filter

Butterworth, drop off gradual

```
Wp = (2*f1)/Fs;
Ws = (2*f2)/Fs;
Rp = 3;    %passband attenuation
Rs = 40;   % stopband attenuation
```

```

[ord,Wn] = buttord(Wp, Ws, Rp, Rs);
[Num, Den] = butter(ord, Wn, 'low');
yf = filter(Num, Den, y);

figure(3);
n = 512;
freqz(Num, Den, n, Fs); % n is evaluation points
% FFT
L = length(yf); % Window
n = pow2(nextpow2(L)); % next power of 2 from length of y
yf_dft = fft(yf,n); % DFT
y_s = fftshift(yf_dft); % Rearrange y values
f = (-n/2:n/2-1)*(Fs/n); % 0 centered frequency range

figure(4)
subplot(2, 1, 1);
plot(t,yf), xlabel('time'), ylabel('Magnitude'),title('Signal in time domain')
subplot(2, 1, 2);
plot(f,abs(y_s)/n), xlabel('Frequency (Hz)'), title('Magnitude response of the
sinusoidal signal')

```

1c High Pass filter

High Pass Elliptic Filter

```

Wp2 = (2*f2)/Fs;
Ws2 = (2*f1)/Fs;
Rp2 = 3;
Rs2 = 40;
[n2,Wn2] = ellipord(Wp2, Ws2, Rp2, Rs2);
[Num2,Den2] = ellip(n2, Rp2, Rs2, Wn2,'high');
yf2 = filter(Num2, Den2, y);

figure(5);
freqz(Num2, Den2, n, Fs);
%
L = length(yf2); % Window
n = pow2(nextpow2(L)); % next power of 2 from length of y
yf2_dft = fft(yf2,n); % DFT
y_s2 = fftshift(yf2_dft); % Rearrange y values
f = (-n/2:n/2-1)*(Fs/n); % 0 centered frequency range

figure(6)
subplot(2, 1, 1);
plot(t,yf2), xlabel('time'), ylabel('Magnitude'),title('Signal in time
domain')
subplot(2, 1, 2);
plot(f,abs(y_s2)/n), xlabel('Frequency (Hz)'), title('Magnitude response of
the sinusoidal signal')

```

1c Bandpass Filter

bandpass elliptic filter design

```

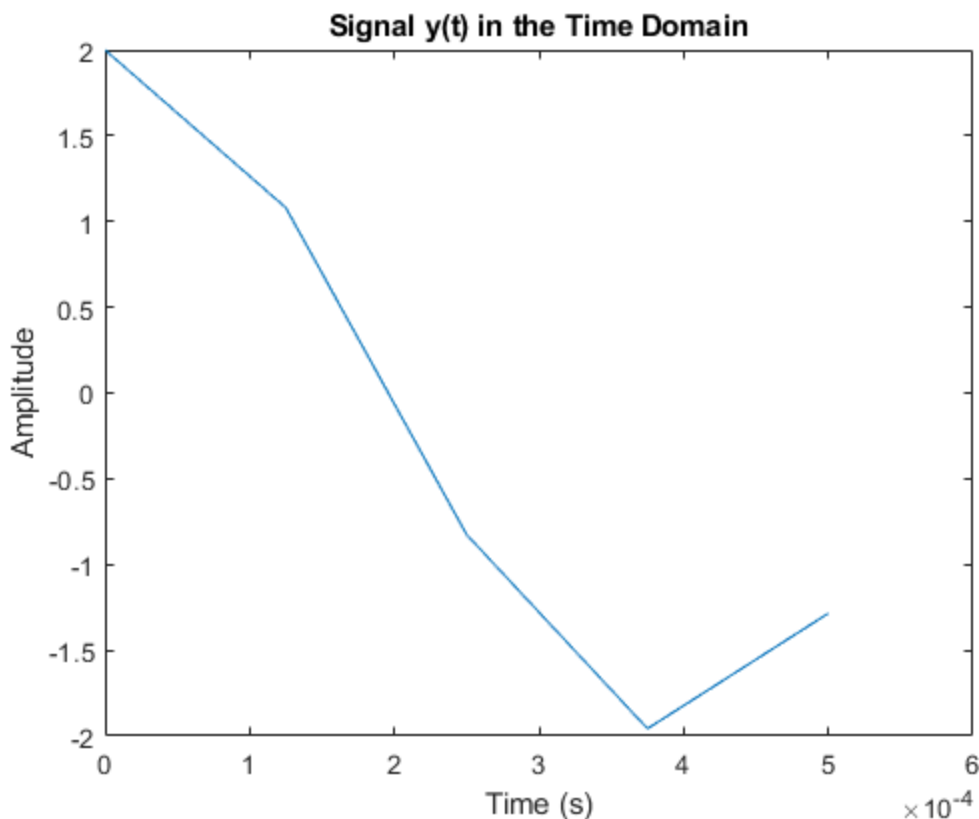
Wpb = (2*[1200 1225])/Fs;
Wsb = (2*[1110 1240])/Fs;
%ws1 < wp1 < wp2 < ws2
Rpb = 3;
Rsb = 40;
[nb, Wnb] = ellipord(Wpb, Wsb, Rpb, Rsb);
[Numb, Denb] = ellip(nb, Rpb, Rsb, Wnb, 'bandpass');
yfb = filter(Numb, Denb, y);

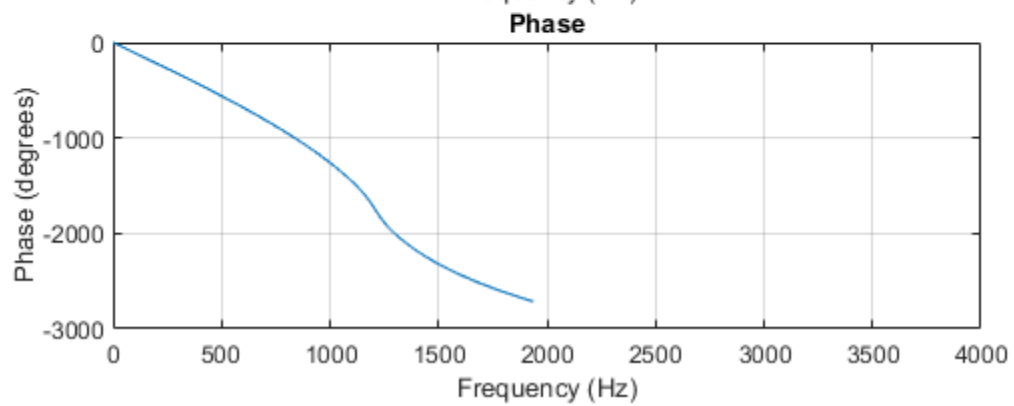
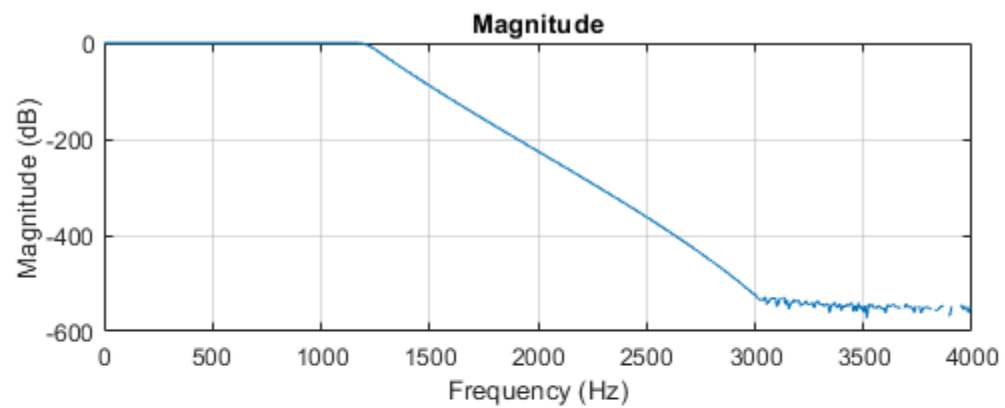
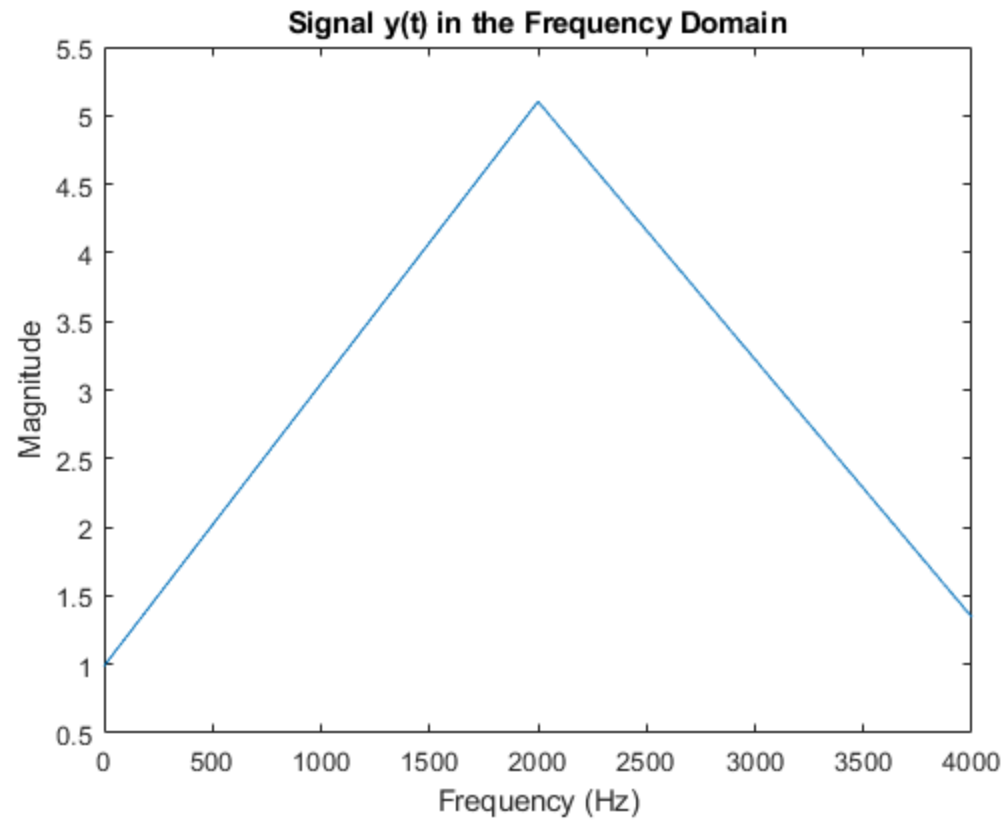
figure(7);
freqz(Numb,Denb, n, Fs);

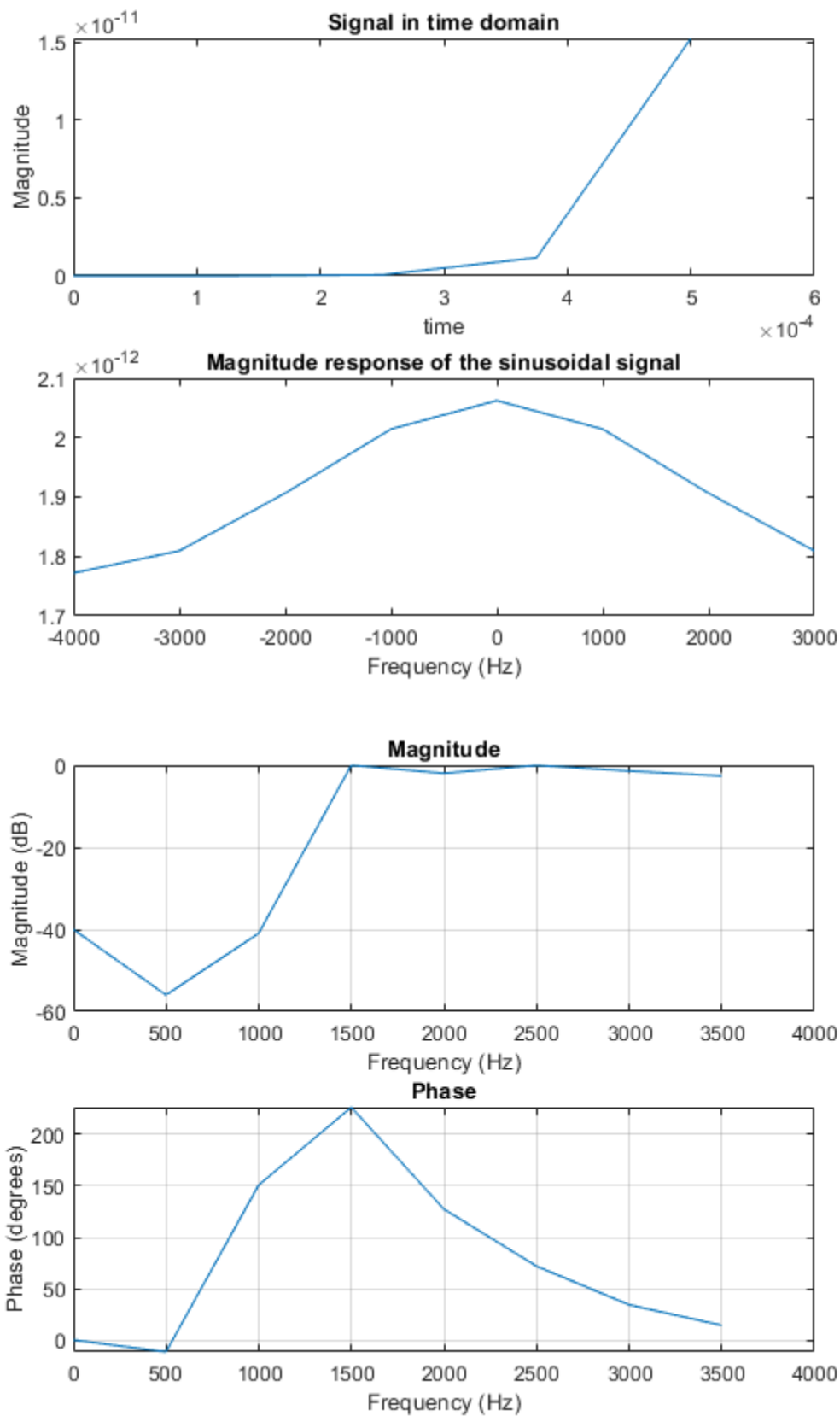
L = length(yfb);           % Window
n = pow2(nextpow2(L));      % next power of 2 from length of y
yfb_dft = fft(yfb,n);      % DFT
y_sb = fftshift(yfb_dft);   % Rearrange y values

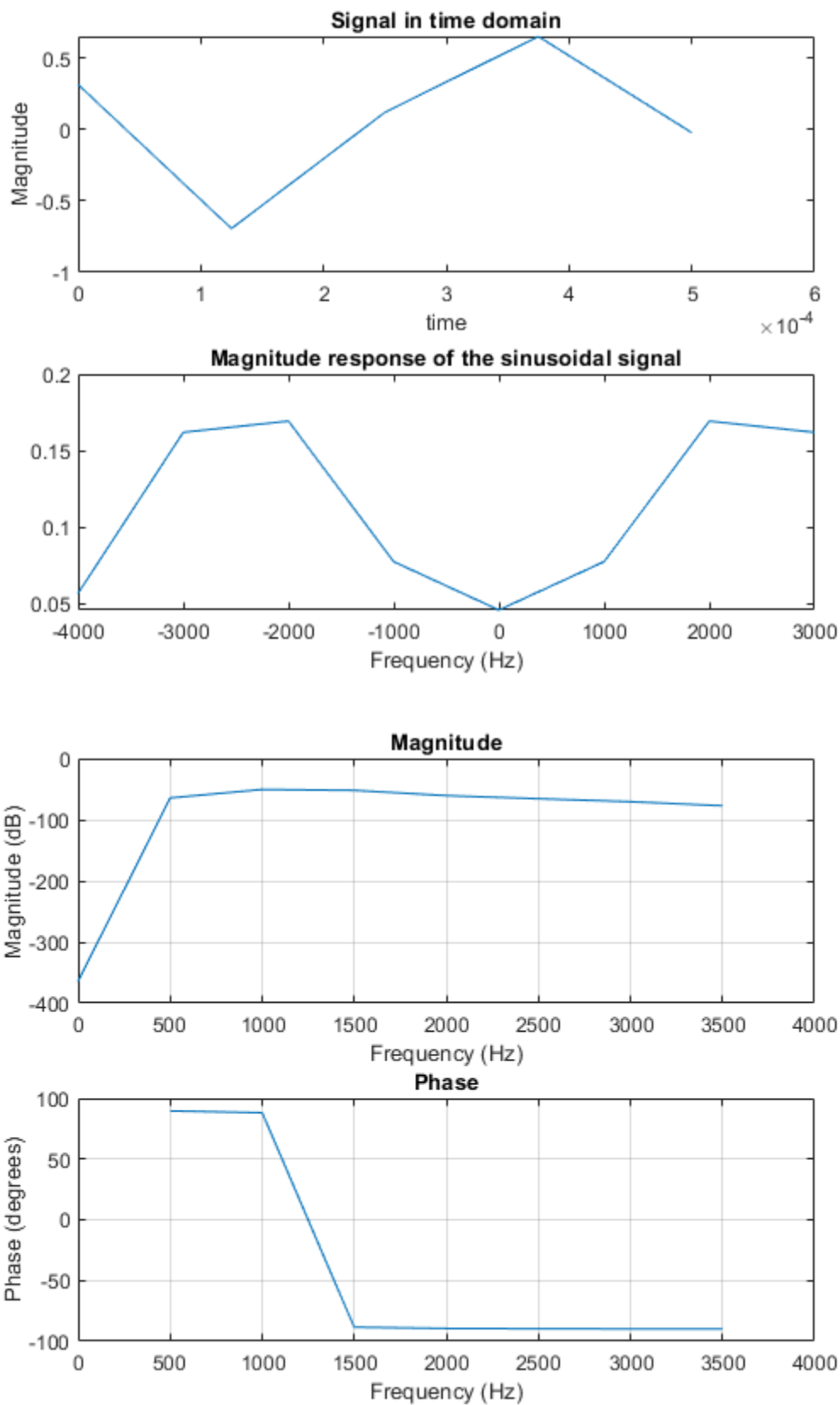
figure(8)
subplot(2, 1, 1);
plot(t,yfb), xlabel('time'), ylabel('Magnititude'),title('Signal in time
domain')
subplot(2, 1, 2);
plot(f,abs(y_sb)/n), xlabel('Frequency (Hz)'), title('Magnitude response of
the sinusoidal signal')

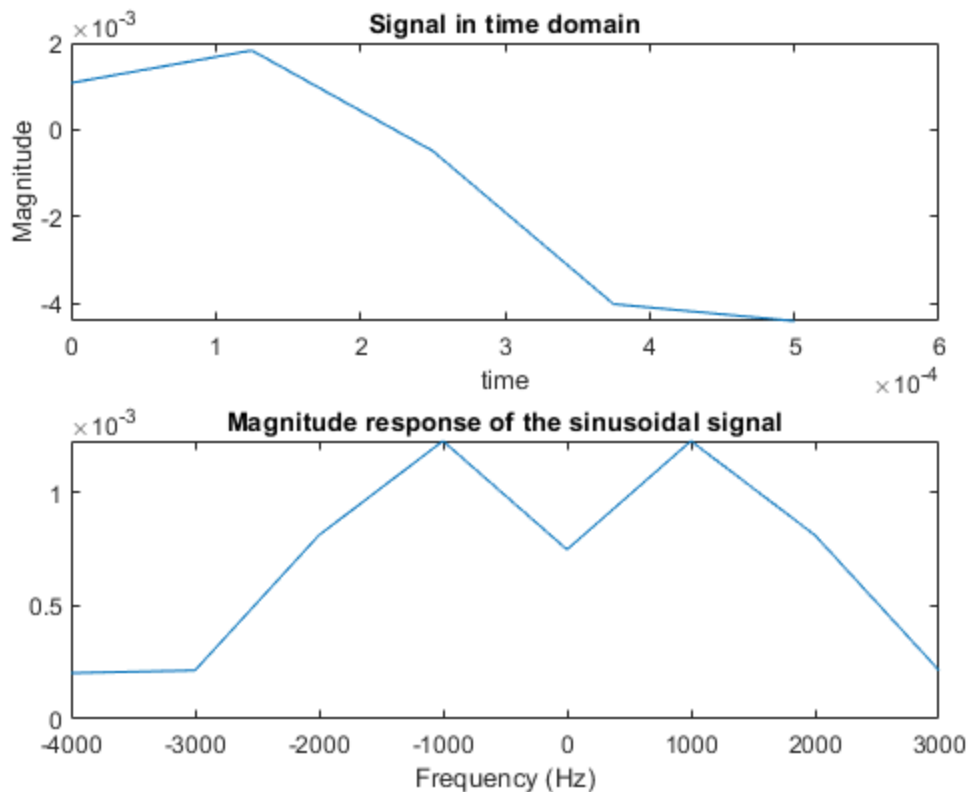
```











Problem 2

```
load SAMPLE_ECG.mat
x = ECG_Data;
Fs = 512;
T = length(x)/Fs;
Ts = 1/Fs;
t = 0:Ts:T-Ts;
```

2a Plot signal vs time and estimate heart rate

```
figure(9)
plot(t,x), xlabel('Time'), ylabel('ECG Signal'), title('ECG Data')
%Estimate Heart Rate (go back and check)
heart_rate = 60 * length(findpeaks(x)) / (t(end) - t(1)); % Assuming
    findpeaks function is available
disp(['Estimated Heart Rate: ' num2str(heart_rate) ' beats per minute']);
```

2b Plot Freq Spectrum

```
L = length(x);
n = pow2(nextpow2(L));
Y = fft(x, n); % Compute the magnitude of the Fourier transform
y_s = fftshift(Y);
```

```
f = (-n/2:n/2-1) * (Fs/n); % Frequency vector based on the sampling rate
figure(10)
plot(f, abs(y_s)/n), xlabel('Frequency (Hz)'), ylabel('Magnitude');
title('Frequency Spectrum of ECG Signal');
```

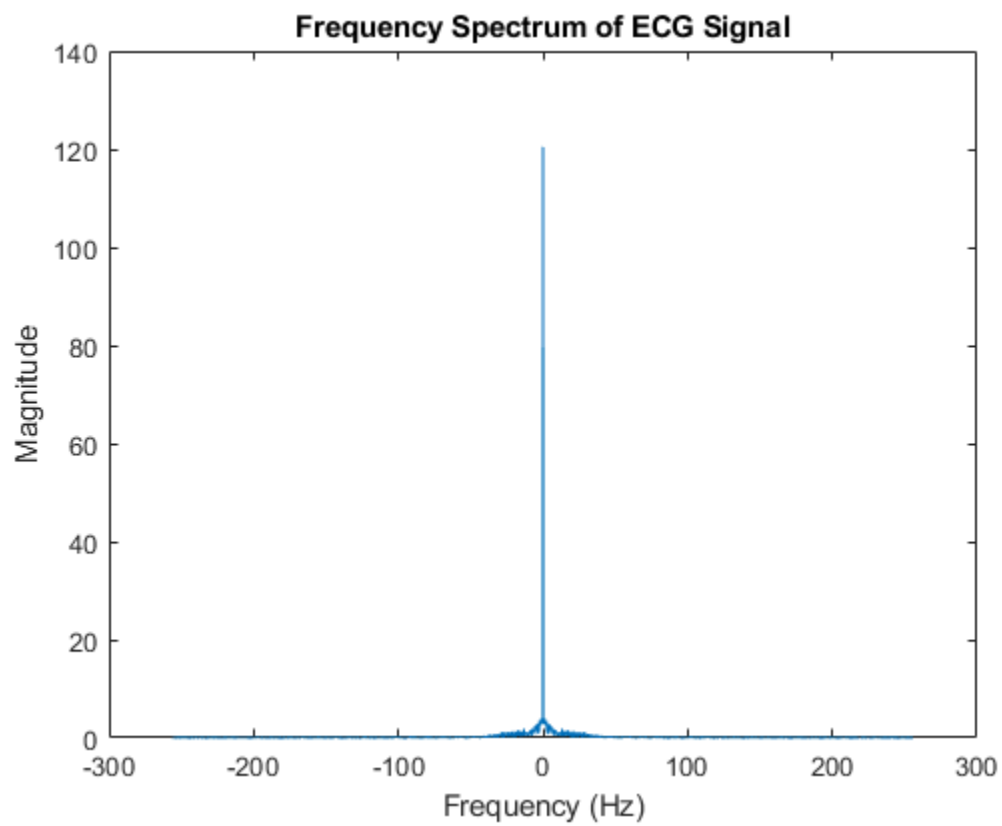
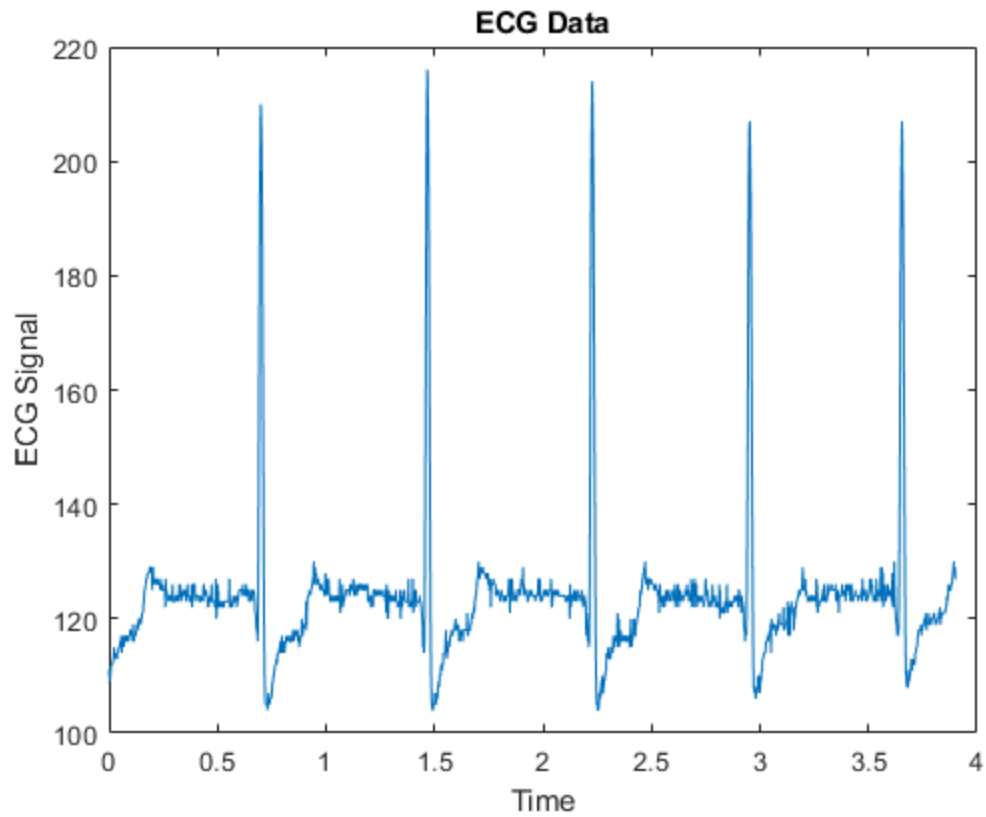
2c Baseline Wander Removal (High Pass filter design)

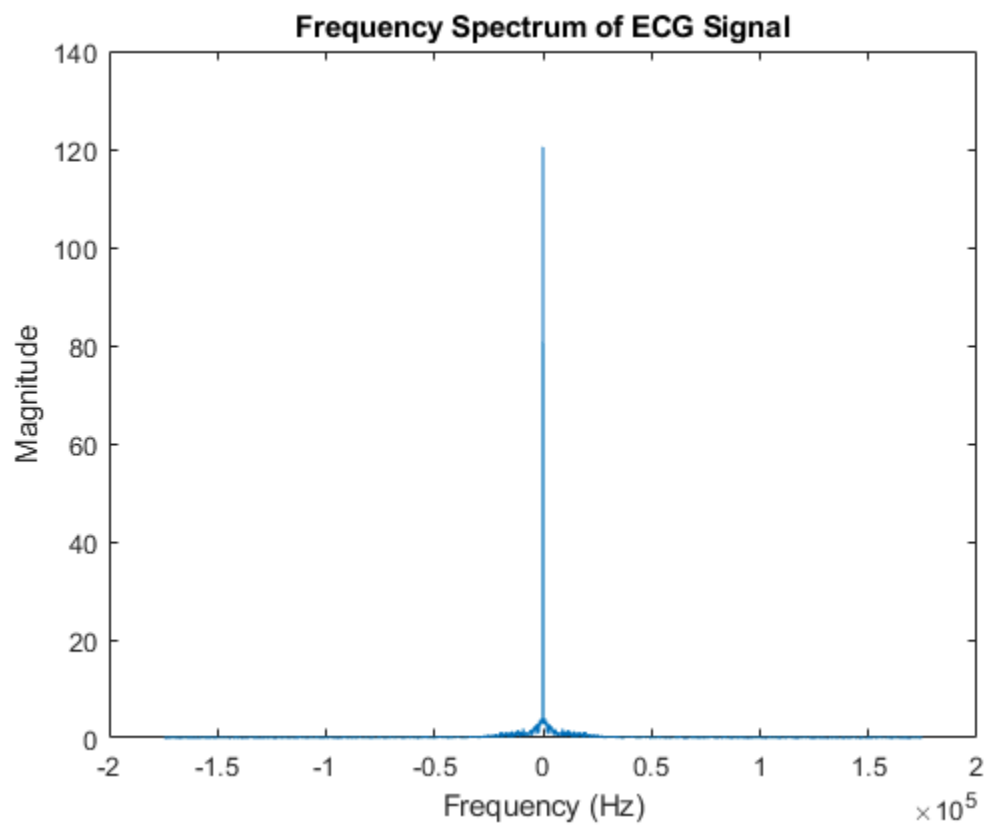
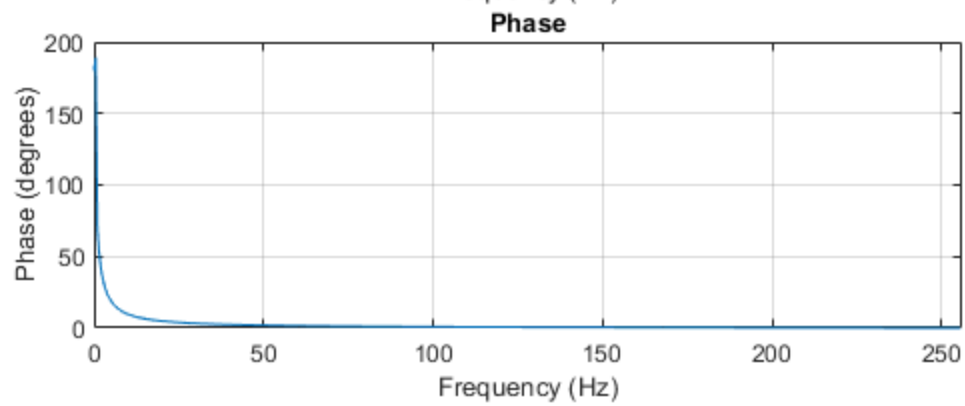
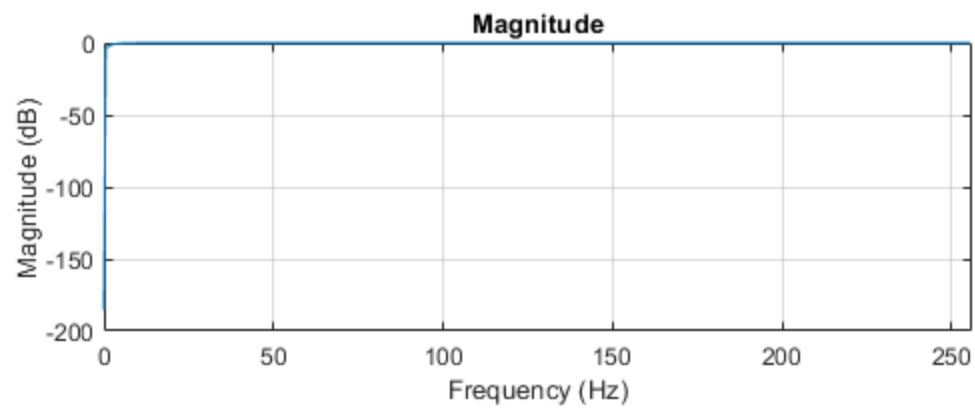
```
fp = 0.5; %passband
fs = 0.2; %stopband
Wp = (2*fp)/Fs;
Ws = (2*fs)/Fs;
Rp = 3;
Rs = 40;
[n2,Wn] = ellipord(Wp, Ws, Rp, Rs);
[Num, Den] = ellip(n2, Rp, Rs, Wn, 'high');
yf = filter(Num, Den, x);

figure(11);
nd = 512;
freqz(Num, Den, nd, Fs);

L = length(x);
n3 = pow2(nextpow2(L));
Y2 = fft(x, n3); % Compute the magnitude of the Fourier transform
y_s2 = fftshift(Y2);
f2 = (-n3/2:n3/2-1) * (Fs/3); % Frequency vector based on the sampling rate
figure(12)
plot(f2, abs(y_s2)/n3), xlabel('Frequency (Hz)'), ylabel('Magnitude');
title('Frequency Spectrum of ECG Signal');
```

Estimated Heart Rate: 5839.7199 beats per minute





Problem 3

```
Fs = 500;
Ts = 1/Fs;
t = 0:Ts:1;
f = 10;      % 10 hz
x = sin(2*pi*f*t); % message

figure(13)
plot(t,x), title(['3a. Message signal with f = ', num2str(f), ' Hz']);
xlabel('Time, t'), ylabel('Amplitude')
```

3b Modulate Signal

```
fc = 100;
y = sin(2*pi*f*t); % message
mod_signal = my_ammod(y,fc,Fs);
```

3c freq spectrum of plots

```
y_psd = periodogram(x);
mod_psd = periodogram(mod_signal);
figure(14);
subplot(2,1,1);
plot(10*log10(y_psd));
title('Frequency Spectrum of Message Signal');
xlabel('Frequency (Hz)'), ylabel('Magnitude');
subplot(2,1,2);
plot(10*log10(mod_psd));
title('Frequency Spectrum of Modulated Signal');
xlabel('Frequency (Hz)'), ylabel('Magnitude');

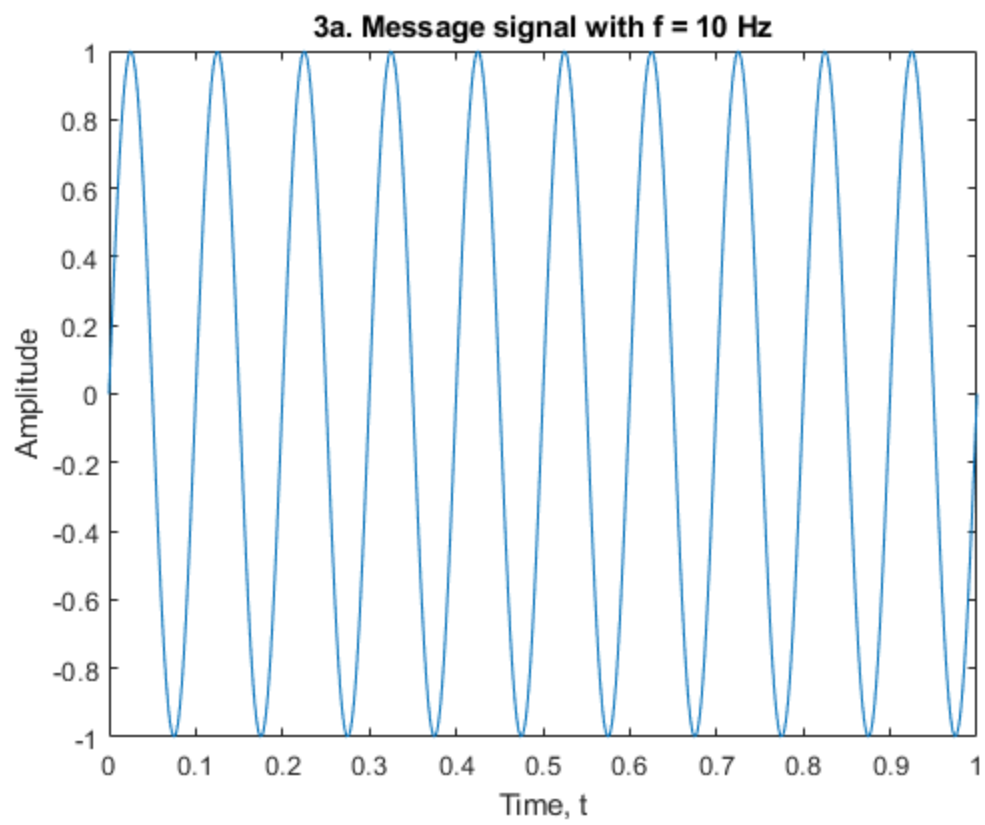
% The difference between the two plots is that the message signal has a
% small peak at 10 Hz whereas the modulated signal has one at 100 Hz
```

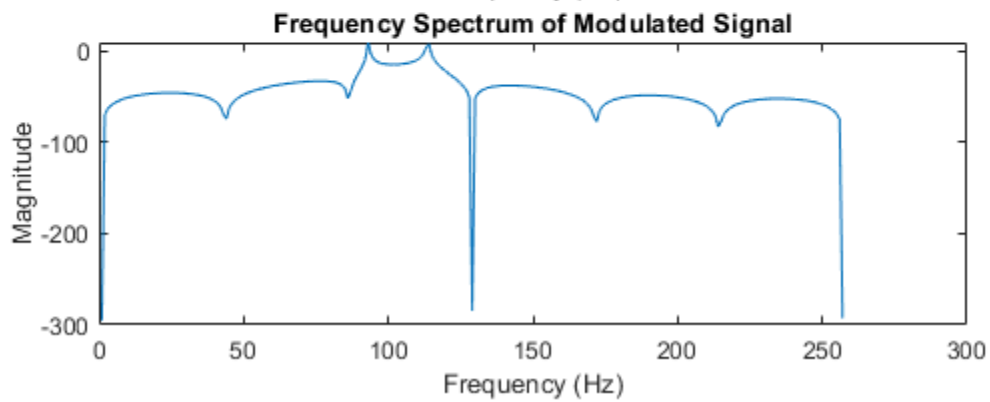
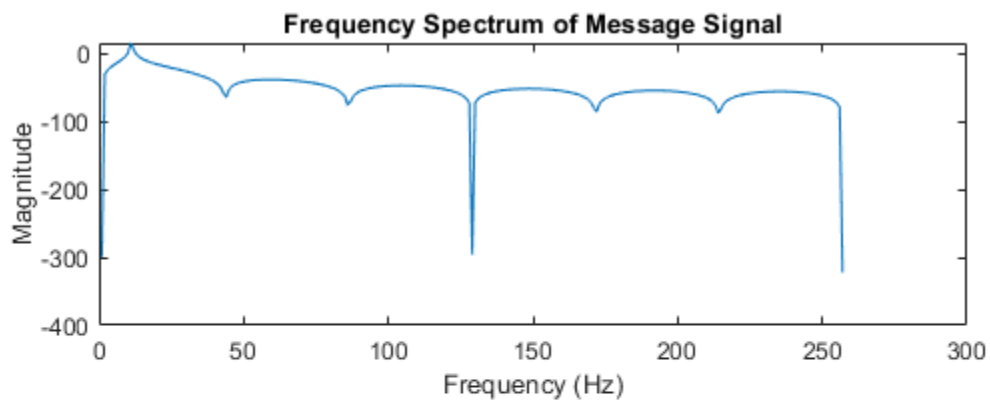
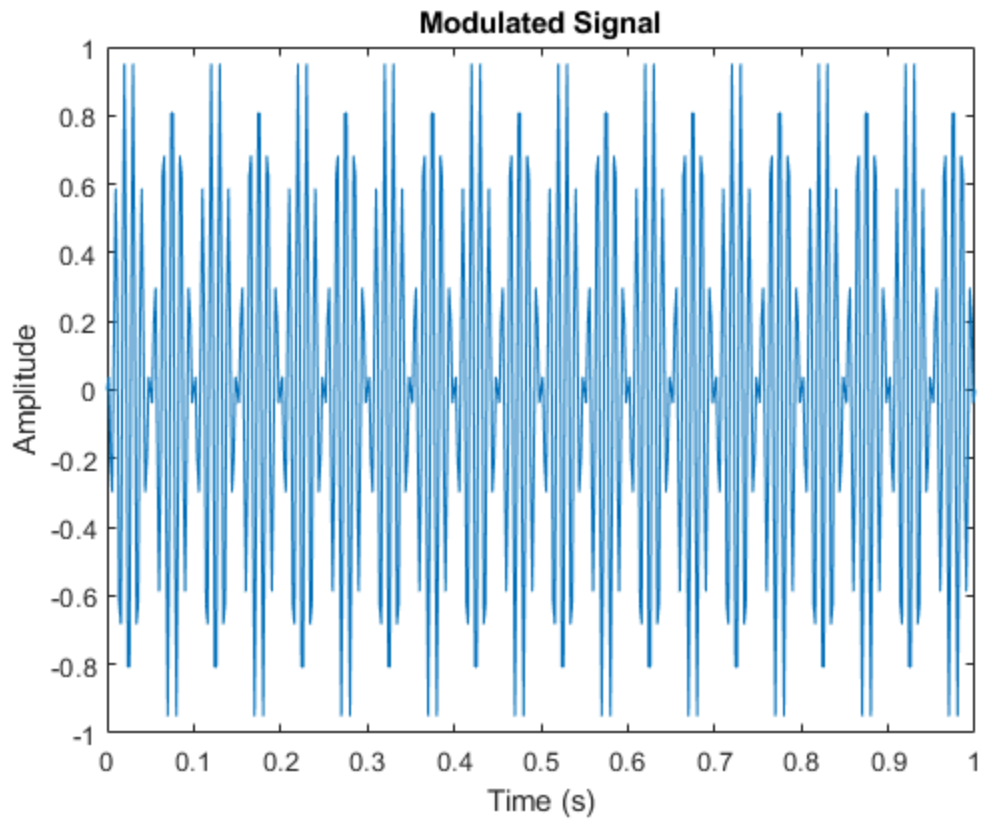
3d Demodulate Signal

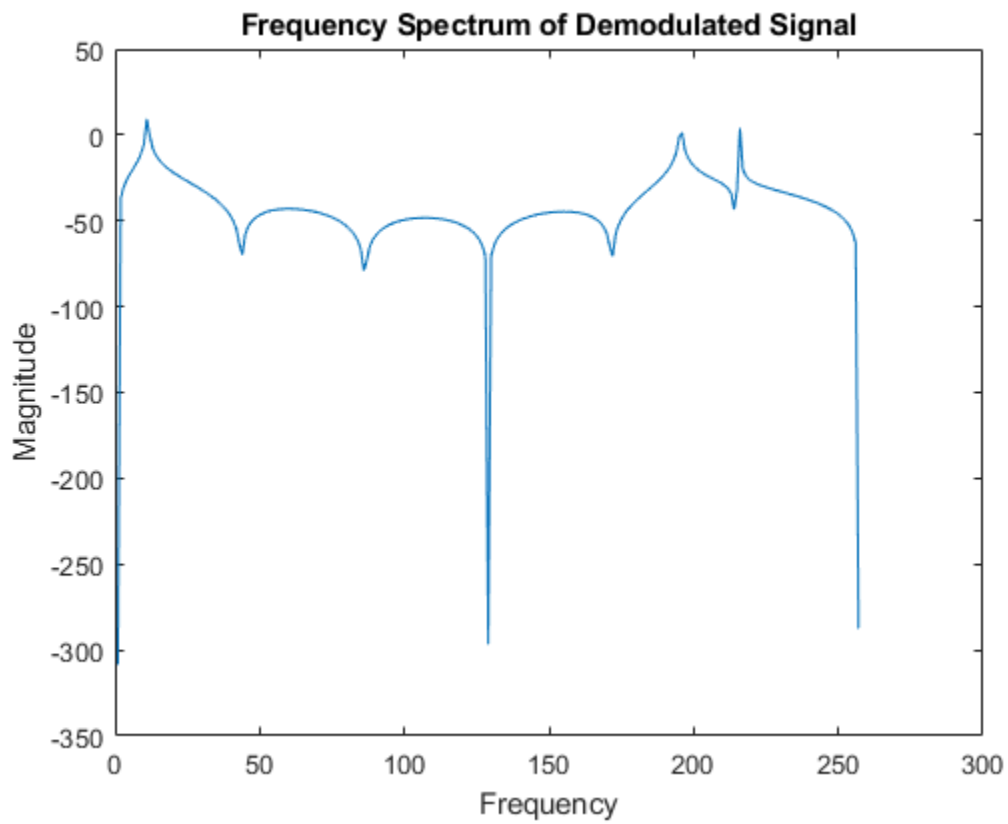
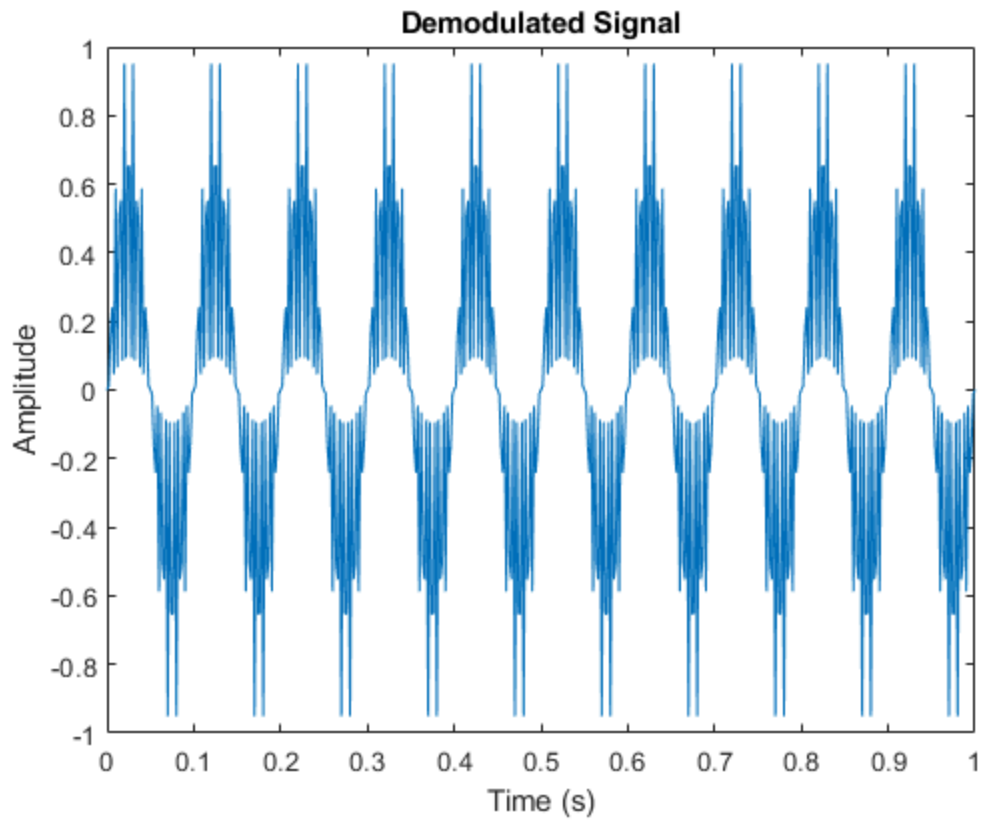
```
demod_signal = my_amdemod(mod_signal,fc,Fs);
% Plot inside function
% Freq spectrum plot
figure(15)
demod_psd = periodogram(demod_signal);
plot(10*log10(demod_psd));
title('Frequency Spectrum of Demodulated Signal');
xlabel('Frequency'), ylabel('Magnitude');

% Compared to the message signal, you can see the message signal a lot more
% clearly than the modulated and demodulated. After demodulating the
% message there is less noise the closer you get to 0 versus the modulated
% one this is because the demodulated signal uses a low pass filter on the
```

% result.







Modulated Signal Function

```
function mod_signal = my_ammod(msg,carFreq, sampFreq)
    t = (0:length(msg) - 1)/sampFreq;
    carSig = cos(2*pi*carFreq*t);
    mod_signal = msg .* carSig;

    % Plot
    figure(16)
    plot(t, mod_signal), xlabel('Time (s)'), ylabel('Amplitude'),
    title('Modulated Signal');
end
```

Demodulated Signal Function

```
function demod_signal = my_amdemod(mod_signal,carFreq, Fs)
    t = (0:length(mod_signal) -1) / Fs;
    carSignal = cos(2*pi*carFreq*t);
    demod_signal = mod_signal .* carSignal;

    figure(17)
    plot(t, demod_signal);
    xlabel('Time (s)'), ylabel('Amplitude'), title('Demodulated Signal');
end
```

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