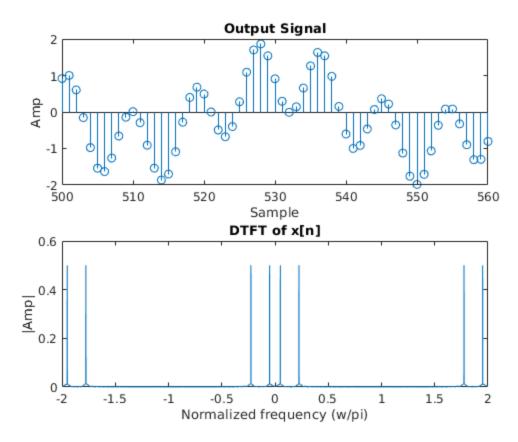
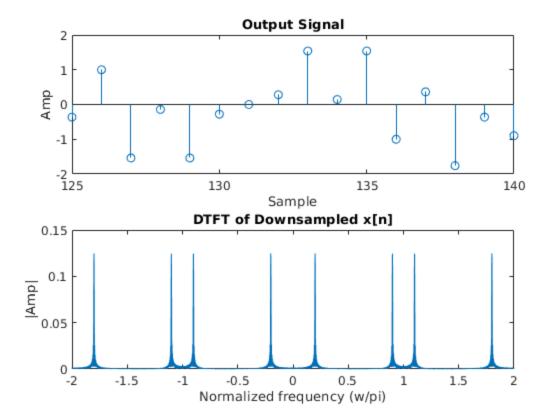
```
%! DSP HW11 #1
%! - Create sampled signal x[n] = sin[2*pi*f1*n*Ts] + sin[2*pi*f2*n*Ts]
%! - Plot samples 500:560 and plot the DTFT
%! - Upsample by 4 and display results
%! Enviorment
clear; close all;
addpath([fileparts(mfilename('fullpath')), '/../../functions']);
%! Variables
Fs = 4000;
Ts
       = 1/Fs;
f1
       = 100;
f2
       = 450;
       = 2048;
       = 0:N-1;
n
n_plot = 500:560;
       = (-2000:2000)*pi/1000;
d_samp = 4;
%! Create Signals
x n = \sin(2*pi*f1*n*Ts) + \sin(2*pi*f2*n*Ts);
x_downsample = downsample(x_n, d_samp);
%! Downsample variables
m = n(1)/4 : n(end)/4;
m plot = 125:140;
w_downsample = w(1:4:end);
%! Take DTFT of singals
         = dtft(x n, n, w) / N;
x_f_{\text{down}} = dtft(x_{\text{downsample}}, m, w) / N;
%! Plot
figure()
subplot(2,1,1)
stem(n_plot, x_n(n_plot))
title('Output Signal')
xlabel('Sample')
ylabel('Amp')
subplot(2,1,2)
plot(w/pi, fftshift(abs(x_f)))
title('DTFT of x[n]')
xlabel('Normalized frequency (w/pi)')
ylabel('|Amp|')
figure()
subplot(2,1,1)
stem(m_plot, x_downsample(m_plot))
xlim([m_plot(1),m_plot(end)])
```

```
title('Output Signal')
xlabel('Sample')
ylabel('Amp')
subplot(2,1,2)
plot(w/pi, fftshift(abs(x_f_down)))
title('DTFT of Downsampled x[n]')
xlabel('Normalized frequency (w/pi)')
ylabel('|Amp|')
```

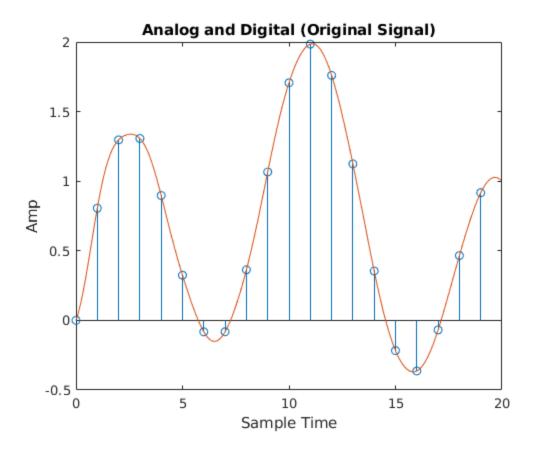


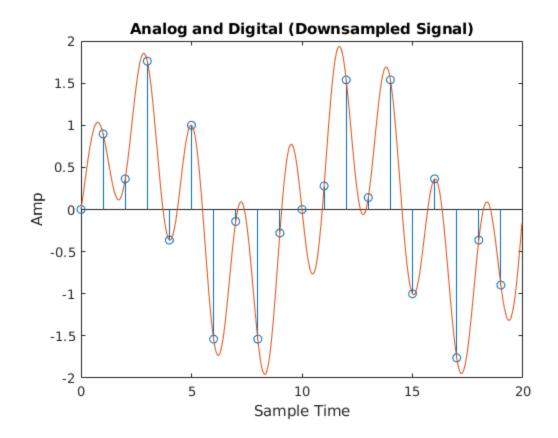


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```
%! DSP HW11 #2
%! - Reconstruct x(t) from x[n]
%! - Reconstruct x(t) from downsampled x[n]
%! - Upsample by 4 and display results
%! Enviorment
clear; close all;
addpath([fileparts(mfilename('fullpath')), '/../functions']);
%! Variables
Fs = 4000;
Ts
      = 1/Fs;
f1
       = 100;
f2
       = 450;
       = 2048;
       = 0:N-1;
d_samp = 4;
t_{delta} = .05;
   = 0:t_delta:N-1;
t
t down = 0:t delta:N/4-1;
%! Create Signals
x_n = \sin(2*pi*f1*n*Ts) + \sin(2*pi*f2*n*Ts);
x_downsample = downsample(x_n, d_samp);
%! Reconstruct original signal using sinc method
x = zeros(N, length(t));
for i=1:N
    x(i, :) = x_n(i) * sinc(t-i+1);
end
xa orig = sum(x);
%! Reconstruct downsampled signal using sinc method
x = zeros(N/4, length(t_down));
for i=1:N/4
    x(i, :) = x_{downsample(i)*sinc(t_{down-i+1)};
xa\_down = sum(x);
%! Plot
%! Plot 20 digital samples
figure()
stem(n(1:20), x_n(1:20))
title('Analog and Digital (Original Signal)')
xlabel('Sample Time')
ylabel('Amp')
hold on
plot(t(1:20/.05), xa_orig(1:20/.05))
hold off
```

```
%! Plot 20 Digital Samples
figure()
stem(n(1:20), x_downsample(1:20))
title('Analog and Digital (Downsampled Signal)')
xlabel('Sample Time')
ylabel('Amp')
hold on
plot(t(1:20/.05), xa_down(1:20/.05))
hold off
```

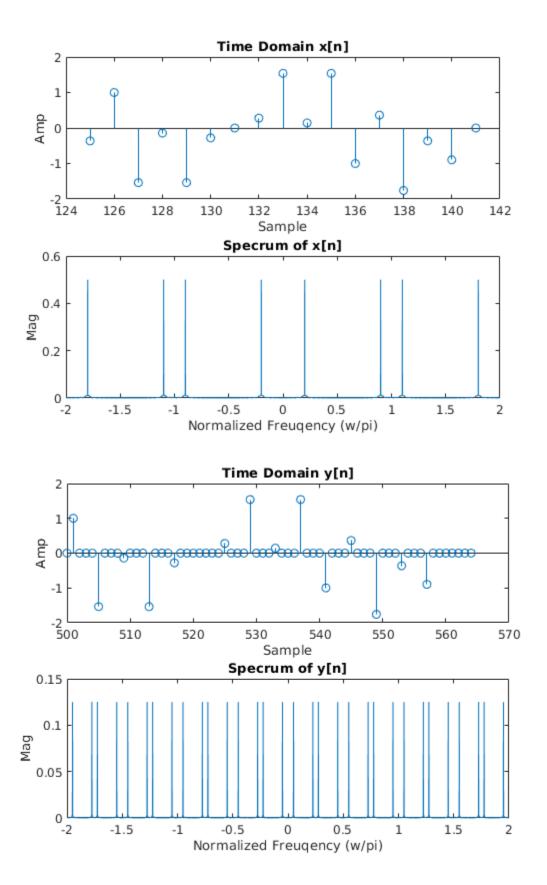


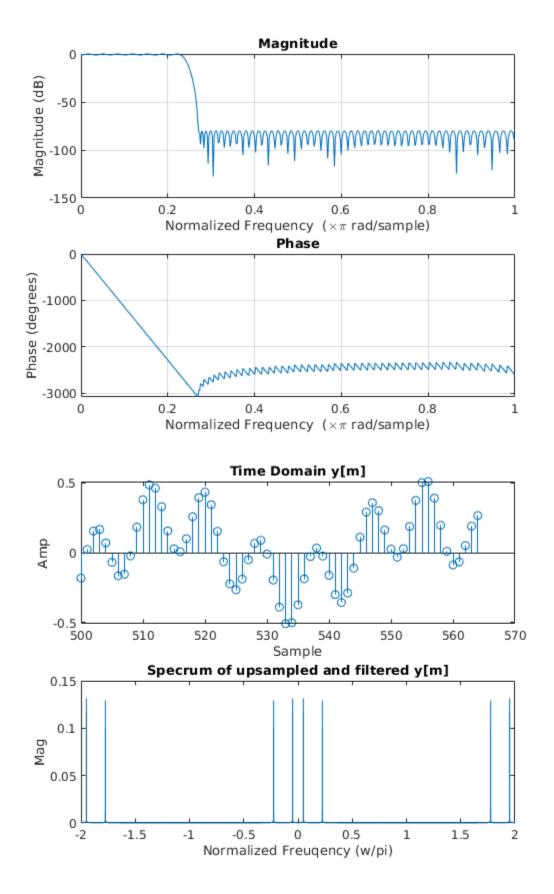


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```
%! DSP HW11 #3
%! - Create x[n] and upsample
%! - Design a filter for the upsampled signal
%! - Display and compare the two singals
%! Enviorment
clear; close all; clc;
addpath([fileparts(mfilename('fullpath')), '/../.functions']);
coeff = load("filters/coeff 3.mat");
%! Variables
       = 1000;
Fs
Ts
       = 1/Fs;
       = 100;
f1
f2
       = 450;
       = 2048;
Ν
       = 0:N-1;
n
up\_samp = 4;
n_plot = 125:141;
        = (-2000:2000)*pi/1000;
%! Create Signals
x_n = \sin(2*pi*f1*n*Ts) + \sin(2*pi*f2*n*Ts);
y_n = upsample(x_n, up_samp);
%! LFP
y_m = filter(coeff.Num, 1, y_n);
%! Take DTFT
x_f = dtft(x_n, n, w) / N;
y f = dtft(y n, 0:N*up samp-1, w) / (N*up samp);
ym_f = dtft(y_m, 0:N*up_samp-1, w) / (N*up_samp);
%! Plot
figure()
subplot(2,1,1)
stem(n_plot, x_n(n_plot))
title('Time Domain x[n]')
xlabel('Sample')
ylabel('Amp')
subplot(2,1,2)
plot(w/pi, abs(x f))
title('Specrum of x[n]')
ylabel('Mag')
xlabel('Normalized Freugency (w/pi)')
figure()
y_plot = n_plot(1)*4:n_plot(end)*4;
subplot(2,1,1)
stem(y_plot, y_n(y_plot))
```

```
title('Time Domain y[n]')
xlabel('Sample')
ylabel('Amp')
subplot(2,1,2)
plot(w/pi, abs(y_f))
title('Specrum of y[n]')
ylabel('Mag')
xlabel('Normalized Freugency (w/pi)')
figure()
freqz(coeff.Num)
figure()
y_plot = n_plot(1)*4:n_plot(end)*4;
subplot(2,1,1)
stem(y_plot, y_m(y_plot))
title('Time Domain y[m]')
xlabel('Sample')
ylabel('Amp')
subplot(2,1,2)
plot(w/pi, abs(ym_f))
title('Specrum of upsampled and filtered y[m]')
ylabel('Mag')
xlabel('Normalized Freugency (w/pi)')
disp(["The new upsampled signal y[m]'s spectrum is the same as the original
 signal in",
    "problem 1. The only differences I have noticed is that a slight phase
 shift in the",
    "time damain signal."])
    "The new upsampled signal y[m]'s spectrum is the same as the origianl
 signal in"
    "problem 1. The only differences I have noticed is that a slight phase
 shift in the"
    "time damain signal."
```







```
%! DSP HW11 #4
%! - Create filters to match dtmf frequencies
   - Filter the signal
%! - Decode the phone number
%! Enviorment
clear; close all; clc;
addpath([fileparts(mfilename('fullpath')), '/../functions']);
        = load("filters/filter697.mat");
f770
       = load("filters/filter770.mat");
f852
     = load("filters/filter852.mat");
       = load("filters/filter941.mat");
f941
f1209
       = load("filters/filter1209.mat");
f1336 = load("filters/filter1336.mat");
f1477 = load("filters/filter1477.mat");
filter_LP = load("filters/filter_LP.mat");
%! Phone Number
phone_num = [1 2 3 4 5 6 7 8 9 0];
%! Call dtmf
[x, Fs] = dtmf(phone_num);
%! Filter the phone signal with each of the filters
   This creates 7 signals each with a single frequency range
y697
      = filter(f697.Num, f697.Den, x);
       = filter(f770.Num, f770.Den, x);
y770
y852
       = filter(f852.Num, f852.Den, x);
y941
       = filter(f941.Num, f941.Den, x);
       = filter(f1209.Num, f1209.Den, x);
y1209
       = filter(f1336.Num, f1336.Den, x);
y1336
       = filter(f1477.Num, f1477.Den, x);
y1477
%! Take power and find if power is above threshold
   If the power in a range of the signal is greater than the threshold
   set that range to 1.
y = zeros(7, length(x));
y697out = filter(filter_LP.Num, filter_LP.Den, y697.^2);
y(1,:) = double(y697out > .3);
y770out = filter(filter_LP.Num, filter_LP.Den, y770.^2);
y(2,:) = double(y770out > .3);
y852out = filter(filter LP.Num, filter LP.Den, y852.^2);
y(3,:) = double(y852out > .3);
y941out = filter(filter LP.Num, filter LP.Den, y941.^2);
y(4,:) = double(y941out > .3);
y1209out = filter(filter_LP.Num, filter_LP.Den, y1209.^2);
y(5,:) = double(y1209out > .3);
y1336out = filter(filter_LP.Num, filter_LP.Den, y1336.^2);
y(6,:) = double(y1336out > .3);
y1477out = filter(filter_LP.Num, filter_LP.Den, y1477.^2);
```

```
y(7,:) = double(y1477out > .3);
%! Downsample the y vectors
% Get one sample per symbol
d = zeros(7, 10);
for i=1:7
    d(i,:) = downsample(y(i,:), 2500, 700);
end
% Array for the phone number pad
phone_num_arr = [1 2 3;
                 4 5 6;
                 7 8 9;
                -1 0 -21;
num=zeros(1,length(x)/2400);
for i=1:10
    idx = find(d(:,i) == 1); % Find idx where the power is > thresh
    row = idx(1);
    col = idx(2)-4;
    num(1,i) = phone_num_arr(row, col);
end
disp('')
disp([' Transmitted Data = ',num2str(num)])
% Plots
figure()
freqz(f697.Num, f697.Den)
title('Frequency Response of BPF for 697Hz')
figure()
freqz(f770.Num, f770.Den)
title('Frequency Response of BPF for 770Hz')
figure()
freqz(f852.Num, f852.Den)
title('Frequency Response of BPF for 852Hz')
figure()
freqz(f941.Num, f941.Den)
title('Frequency Response of BPF for 941Hz')
figure()
freqz(f1209.Num, f1209.Den)
title('Frequency Response of BPF for 1209Hz')
figure()
freqz(f1336.Num, f1336.Den)
title('Frequency Response of BPF for 1336Hz')
figure()
freqz(f1477.Num, f1477.Den)
title('Frequency Response of BPF for 1477Hz')
figure()
subplot(7,1,1)
plot(y697)
subplot(7,1,2)
plot(y770)
subplot(7,1,3)
```

```
plot(y852)
subplot(7,1,4)
plot(y941)
subplot(7,1,5)
plot(y1209)
subplot(7,1,6)
plot(y1336)
subplot(7,1,7)
plot(y1477)
xlabel('Samples')

figure()
for i=1:7
    subplot(7,1,i)
    plot(y(i,:))
end
```

Transmitted Data = 1 2 3 4 5 6 7 8 9 0

