## Question 1

a) Matlab code for findpeaks.m:

```
function [peaks,freqs]=findpeaks(Xwdb,maxPeaks,fs,win,N)
% peaks = a vector containing the peak magnitude estimates (linear) using
          parabolic interpolation in order from largest to smallest peak.
% freqs = a vector containing the frequency estimates (Hz) corresponding
          to the peaks defined above
% Xwdb = DFT magnitude (in dB scale) vector of a windowed signal.
          NOTE that it may contain
%
          only negative-frequency (length < N/2+1), positive-frequency
          (length = N/2+1), or all (length = N) bins of the FFT.
% maxPeaks = the number of peaks we are looking for
% fs = sampling frequency in Hz
% win = window used to obtain Xwdb (assumed zero phase)
% N = NFFT, the number of points used in the FFT creating Xwdb
if (length(Xwdb) == N)
    Xwdb = Xwdb(1:floor(N/2));
end
%-- Find all peaks (magnitudes and indices) by comparing each point of ---%
%-- magnitude spectrum with its two neighbors ---%
allPeaks = [];
for i=2:length(Xwdb)-1
    if (Xwdb(i) > Xwdb(i-1) && Xwdb(i) > Xwdb(i+1))
        newPeak = [Xwdb(i); i];
        allPeaks = [allPeaks newPeak];
    end
end
%-- Order from largest to smallest magnitude, keep only maxPeaks of them --%
[Mags, Inds] = sort(allPeaks(1,:), 2, 'descend');
peak_matrix = allPeaks(:,Inds);
%-- Resize to only include the first maxPeaks columns --%
peak_matrix(:, 4:size(peak_matrix,2)) = [];
```

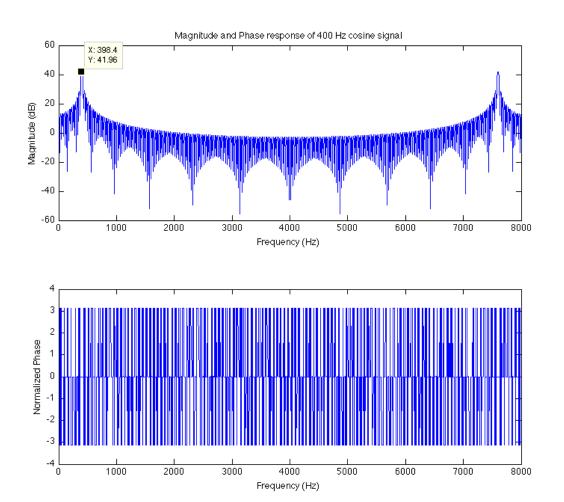
```
freq_indx = [];
  db_peaks = [];
  %-- Do parabolic interpolation in dB magnitude to find more accurate peak --%
  %-- and frequency estimates --%
  for i=1:min(length(peak_matrix), maxPeaks)
       idx=peak_matrix(2,i);
      %parabolic interpolation
       a=Xwdb(idx-1);
      b=Xwdb(idx);
       c=Xwdb(idx+1);
       [p,y,a] = qint(a,b,c);
      db_peaks = [db_peaks, y];
       freq_indx = [freq_indx, (idx + p)];
   end
  %-- Return linear amplitude and frequency in Hz --%
  % NOTE that we must use knowledge of the window to normalize amplitude here
  \% if we have a TD cosine of amplitude 0.6, this output should be 0.6
  win_max = max(win);
  lin_peaks = db2mag(db_peaks);
  peak_max = max(lin_peaks);
  peaks = win_max * (lin_peaks / peak_max);
  freqs = (freq_indx-1) * (fs / N);
b) Matlab code for p1b.m:
  fs = 8000;
  T = 1/fs;
  f = 400:
  n = (-127: 127);
  sinusoid = cos(2 * pi * f * n * T);
  zpzp_Xw = fft(zpzpwin(cos(2*pi*400/8000*(-127:127)'),boxcar(255),2048));
  subplot(211);
  plot([0: fs/2048 : fs - 1], 20*log10(abs(zpzp_Xw)));
  title('Magnitude and Phase response of 400 Hz cosine signal');
  ylabel('Magnitude (dB)');
  xlabel('Frequency (Hz)');
```

```
subplot(212);
plot([0: fs/2048 : fs - 1], angle(zpzp_Xw));
ylabel('Normalized Phase');
xlabel('Frequency (Hz)');

[peaks, freqs] = findpeaks(20*log10(abs(zpzp_Xw)),2,8000,boxcar(255),2048);
peaks
freqs

The output of the above code is:

peaks = 1.0000     0.2178
freqs = 400.0074     445.5445
and the plot looks like*:
```

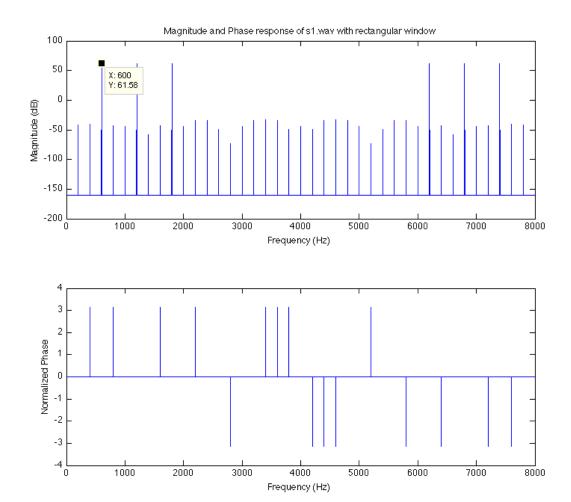


<sup>\*</sup>Note that the graphical indication of the peak is found simply by the maximum sample, not through the parabolic interpolation, so it is not as accurate as the value returned in freqs.

c) Matlab code for p1c.m:

```
[x fs] = wavread('s1.wav');
N = length(x);
Xlin = fft(x);
eps = .00000001;
Xwdb = 20*log10(abs(Xlin)+eps);
subplot(211);
plot([0: fs/N : fs - 1], Xwdb);
title('Magnitude and Phase response of 400 Hz cosine signal');
ylabel('Magnitude (dB)');
xlabel('Frequency (Hz)');
subplot(212);
plot([0: fs/N : fs - 1], angle(Xlin));
ylabel('Normalized Phase');
xlabel('Frequency (Hz)');
[peaks, freqs] = findpeaks(Xwdb,4,fs,boxcar(length(x)),length(x));
peaks
freqs
The output of the above code is:
peaks = 1.0000
                  1.0000
                            1.0000
freqs = 1200
                     600
                                 1800
```

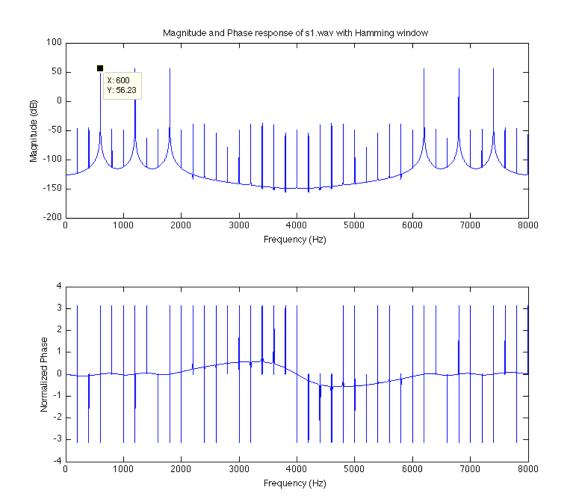
## and the plot looks like:



d) Matlab code for p1d.m:

```
[x fs] = wavread('s1.wav');
N = length(x);
ham = hamming(length(x));
x_win = ham .* x;
Xlin = fft(x_win);
eps = .00000001;
Xwdb = 20*log10(abs(Xlin)+eps);
subplot(211);
plot([0: fs/N : fs - 1], Xwdb);
title('Magnitude and Phase response of s1.wav with Hamming window');
ylabel('Magnitude (dB)');
xlabel('Frequency (Hz)');
subplot(212);
plot([0: fs/N : fs - 1], angle(Xlin));
ylabel('Normalized Phase');
xlabel('Frequency (Hz)');
[peaks, freqs] = findpeaks(Xwdb, 4, fs, ham, length(Xwdb));
peaks
freqs
The output of the above code is:
peaks = 1.0000
                  1.0000
                             1.0000
freqs = 1.0e+03 *
                       0.6000
                                  1.8000
             1.2000
```

and the plot looks like:

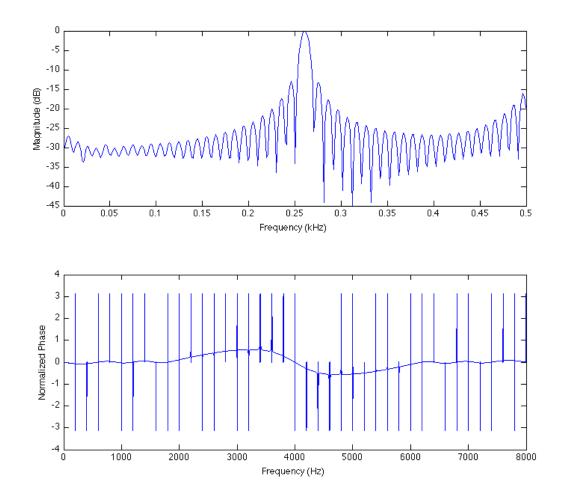


## Question 2

- a) (nothing to show)
- b) (nothing to show)
- c) The output of oboeanal is:

```
Read oboe.ff.C4B4.wav, fs = 44100.000000, nbits = 16, length = 29381, samples = 0.7 sec 
 Estimated pitch = 260.710019 Hz 
 Nw = 2048 
 Nw = 4096 
 Nw = 8192
```

and the plot in Figure 1 looks like:



The frequency value at the peak magnitude lobe agrees exactly with the Estimated pitch of 260.71.

- d) All three windows show an ability to resolve the peaks, although the results clearly get better and better as the length of the window grows. By this I mean that the side lobes drop lower down and the noise floor gets lower, especially towards the higher frequencies.
- e) The trend is that larger values of K produce better, more resolvable results. Specifically, when K=1 the plot is very unclear, and the graph seems to indicate that the magnitude peaks are in the wrong order: it shows the third peak from the left as higher than the first and second peaks, as all the other graphs do.