

Homework Set 2

Due by 12:00 noon on Tuesday, Oct. 16, 2007 in 16-352.

1 Introduction

Here we will look at power spectra in MATLAB, and some issues associated with discrete-time Fourier analysis. You should save your MATLAB code in an m-file, which will make it easy to modify and reuse it, and submit your m-file along with your plots.

Remember, when in doubt about any of the MATLAB commands given below, use the `help` command to get info about how to use each one – the Windows installation of MATLAB will have this same info and more accessible via the Help menu.

Matlab command for computing power spectra is `pwelch`. To get a feeling of what `pwelch` does, here is a diagnostic problem and answer to the problem associated with the command. **Try problem 1 in the problem section first**, and then come back to the diagnostic problem. It is also beneficial to take a look at the Matlab help page for `pwelch` for its input and output arguments and other detailed information as you read through this section.

1.1 Diagnostic problem

An important feature of `pwelch` is that it always correctly normalizes the total power of the PSD, but—depending on the parameters you use—you can get quite different PSD shapes for the same signal. For example, the default parameters give a lot of *spectral leakage*. To see a low-leakage spectrum, try running `pwelch` with the `nfft` and `window` parameters satisfying the following conditions:

- the sinusoid frequency f_{SIN} is an integer multiple of the quantity f_{samp}/N_{fft}
- `window` is the same length as `nfft`.

Can you suggest what causes spectral leakage? Plot the low-leakage PSD on the same axes as your previous linear and log plots (use `hold` after plotting one waveform to freeze a figure, before plotting a second). Considering the relative magnitudes, how much does the leakage matter?

1.2 Answer

(See Figure on next page.)

In order to get a low-leakage PSD, the distribution of FFT points along the frequency axis must be such that one of the points exactly coincides with the frequency of the sinusoid. Leakage happens when signal power exists at frequencies *between those represented by FFT points*. Since, in practice, this happens anytime you don't know precisely what the input function is (and if you do, there's no point in measuring its spectrum), leakage is seen. Being aware of it, and knowing how it can be reduced is important.

There are N_{fft} points in a PSD computed by `pwelch`, which are distributed evenly through the region of the spectrum between $-f_{samp}/2$, and $f_{samp}/2$ (the positive and negative frequencies are then combined for a one-sided PSD, since the FFT of real signals is symmetric) therefore N_{fft} must be chosen to obey

$$f_{SIN} = k \frac{f_{samp}}{N_{fft}}, \quad [k = 1, 2, 3, \dots].$$

In other words, the frequency of the sinusoid needs to be an integer multiple of `fsamp/nfft`. Of course, `nfft` should be large enough to give a decently tight spacing of points in the PSD.

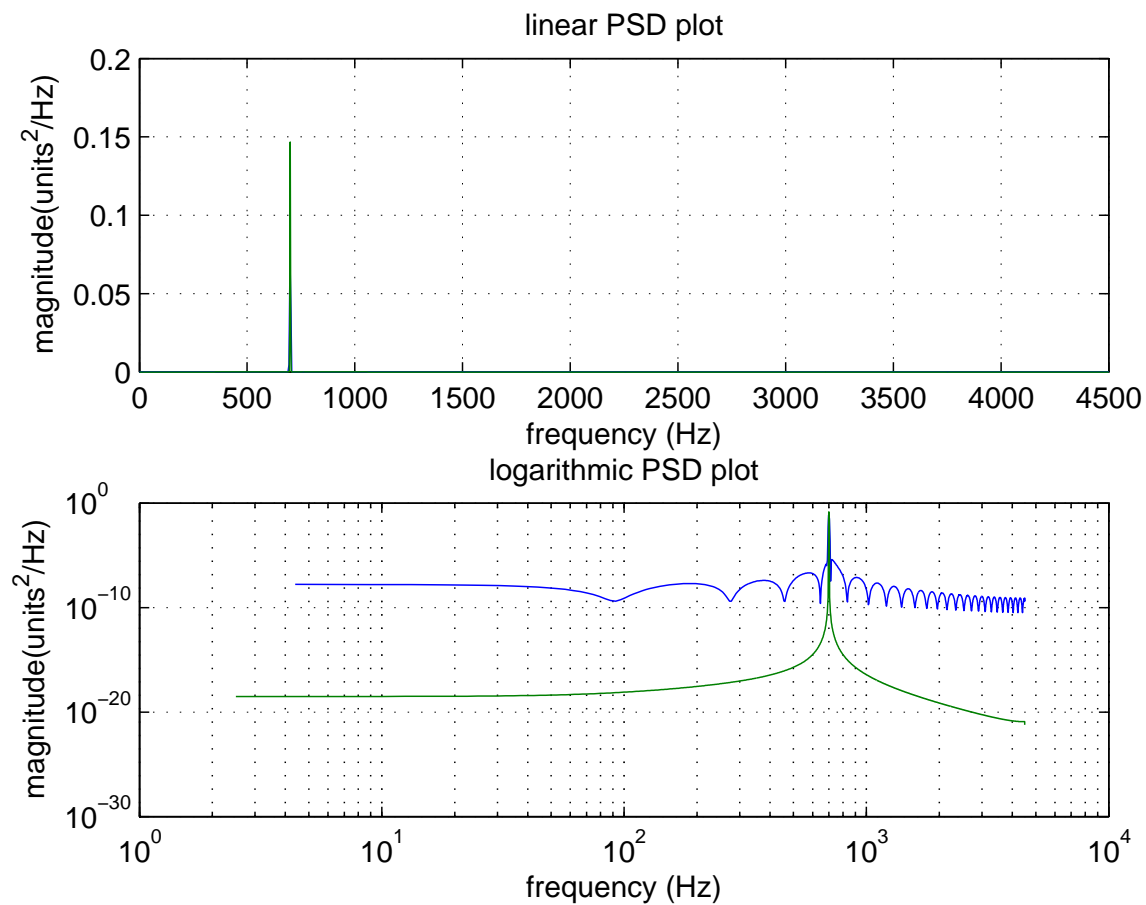


Figure 1: Linear and logarithmic plots of a PSD for a 700Hz sine wave sampled at 9KHz calculated with `pwelch(signal,3600,[],3600,9000)`. Note that the log plot is crucial for properly visualizing the leakage that `pwelch` causes by default. Since the leaked signal power in the blue curve is five or more orders of magnitude smaller than the sinusoid, it can be considered to be zero for most purposes, and therefore doesn't much matter. There may be times when being aware of such leakage is important.

2 Problems

1. Use MATLAB to generate a time vector $y(t)$, approximately one second long, with a sampling frequency f_{samp} of a few kHz (the syntax “`vector=start_value:increment:end_value;`” will be useful here — `increment` is the time between your samples or $1/f_{\text{samp}}$). Then create a sinusoid based on that time vector — choose a frequency f_{SIN} of a few hundred Hz:

$$y = \sin(\omega_{\text{SIN}}t) = \sin(2\pi f_{\text{SIN}}t)$$

Calculate the PSD of your sinusoid using the `pwelch`¹ command, and plot it on a linear scale and a logarithmic scale; use the `plot` and `loglog` commands, respectively. What is the significance of the highest and lowest frequencies that appear on the plot?

2. Recall that one of the consequences of Parseval’s theorem is the following relationship between a time-domain signal $f(x)$ and its PSD $F(\omega)$:

$$\langle f(x)^2 \rangle = \int_0^\infty F(\omega) d\omega.$$

Verify that this is the case for the sine wave you’ve been using by computing its mean-square value in the time domain, and the integral of its PSD in the frequency domain (MATLAB’s `var` and `sum` functions will be useful here). Remember that you’re effectively calculating an area, and make sure that units match up: `pwelch` gives you the PSD in units^2/Hz , while the integral of your PSD needs to be equal to a mean-square value (units^2).

3. Now use the `randn` command to generate a noise signal with the same length as your time vector. Calculate its PSD with the `pwelch` parameters of your choice, and plot it on a new log plot. To observe the benefits of averaging, generate a noise signal that is $10\times$ longer in duration, calculate its PSD, using the same `pwelch` parameters, and plot its spectral density on the same plot.
4. Take the sinusoid from problem 1, but with its amplitude reduced tenfold, and the first (short duration) noise signal from problem 3, and add them together. Look at a section of the summed waveform in the time domain - can you find the sine wave at all? Now plot the PSD of the combined signal - can you find the sine wave peak in the noise? What can you do to get it to resolve more clearly? (Problem 3 should provide a clue.) Plot your result on the same axes.
5. (**BONUS**) Finally, take the original sinusoid from (a), compute its Discrete Fourier Transform using the `fft` command, and plot its magnitude (an FFT is complex-valued) on a semi-log plot (see the `semilogy` command). Compare this to the PSD of the sine wave given by `pwelch`, and comment on the FFT’s features: Why two peaks? What is the meaning of its x-axis values? Why are the values along the y-axis so large?

¹The syntax is `[PSDvect, freqvect]=pwelch(signal>window,n_overlap,nfft,f_samp)` and only requires you to supply a `signal` vector and a number for `f_samp` to properly scale and calculate the frequencies for the PSD – i.e. `pwelch(signal,[],[],[],fsamp)` will get you a result. The result is stored in the two vectors before the `=` sign. Use MATLAB’s `help` to find out what the other parameters do. For parameters that you leave out by entering “[]”, MATLAB uses its defaults (also found in help).