Discrete Fourier Transform

Lab 8, DSP

Objective

Students should understand the Discrete Fourier Transform, and be able to use the FFT algorithm in Matlab for frequency analysis of data.

Theoretical aspects

Exercises

- 1. Generate the signal $x[n] = 0.5\sin(2\pi f_1 n) + 0.7\sin(2\pi f_1 n) + noise$, where $f_1 = 0.1$, $f_2 = 0.15$, $n = \overline{1:10000}$ and noise is AWGN with variance $sigma^2 = 0.1$.
 - a. Plot the signal, in the top half of a sigure (use subplot()).
 - b. Display the magnitude spectrum of the signal x[n] from 1, two-sided (ω going from $\omega = -\pi$ to $\omega = \pi$), in the bottom half of the figure (use subplot()).
- 2. Write a script file to implement a live spectrum analyzer.
 - a. Read 8192 samples from the system's audio input device (use audiorecord())
 - b. Plot the signal and its magnitude spectrum as in the previous exercise. Plot only the positive side of the magnitude spectrum (ω cgoing from $\omega=0$ to $\omega=\pi$) plot only
 - c. Go back to a. and repeat forever (you can stop it with Ctrl-C)
- 3. Write a function p = pitch(x) which returns the dominant frequency in the input signal x (the dominant frequency is the frequency for which the magnitude spectrum is maximum). Call the function from the iterations in the previous exercise, and display the dominant frequency in Matlab command line (use disp()).

Final questions

1. TBD