Due:

Problem 1: Using filter and difference equation

Matlab provides the filter function which is a direct form II transposed implementation of the standard difference equation. Such function saves coding time and helps us focus more on system behavior.

- a) Re-do problem 3 of class work 2 and implement $y(n) = \frac{x(n) + x(n-1) + x(n-2)}{3}$ using the 'filter' command to filter out the high frequency components of the ECG signal.
- b) Instead of using 'filter' command, convolute the ECG signal with h=[1/3, 1/3, 1/3] using 'conv' function. Compare the result with that of (a).
- c) Write your own convolution function and re-do (b) with yours.

Problem 2: basic about sinusoids and sampling rate

Let the sampling rate Fs be 50 Hz, and the time array t varies from 0 to 1 second.

- a) Create and plot a sine signal $y_1(t) = \sin(2\pi f_1 t)$, where $f_1 = 10 \text{ Hz}$
- b) Create and plot a sine signal $y_2(t) = \sin(2\pi f_2 t)$, where $f_2 = 60 \ Hz$
- c) Can you differentiate between $y_1(t)$ and $y_2(t)$ plot? Why?

Problem 3: sound

Let the sampling rate Fs be 50 kHz, and the time t varies from 0 to 1 second.

- a) Create and plot a 2 kHz signal $y_1(t) = \sin(2\pi f_1 t)$ using Matlab. Let the horizontal axis vary from 0 to 0.01 second. Using the command 'sound(signal, sampling rate)' to listen to the tone.
- b) Create, plot and listen to a sine signal $y_2(t) = \sin(2\pi f_2 t)$, where $f_2 = 6$ kHz. How the sound of y_2 in comparison to the sound of y_1 ?
- c) Create, plot and listen to a sine signal $y_3(t) = \sin(2\pi f_3 t)$, where $f_3 = 25$ kHz. Can you still hear the sound this time?