EE110B Lab 6

From the lectures, we see the importance of lowpass filters. In this lab exercise, you will learn and test a simple technique to design a digital lowpass filter. Let h[n] be the impulse response of an ideal lowpass filter with the frequency response

$$H(f) = \begin{cases} 1, & |f| \le 0.2\\ 0, & 0.2 < |f| \le 0.5 \end{cases}$$

where the cutoff frequency is chosen to be 0.2.

- 1) First, determine the analytical form of h[n] by performing the inverse DTFT of H(f).
- 2) Second, compute $g[n] = h[n n_0]w[n]$ where

$$w[n] = \begin{cases} n, & 0 \le n \le n_0 \\ 2n_0 - n, & n_0 \le n \le 2n_0 \end{cases}$$

Unlike h[n], g[n] is a finite casual impulse response.

- 3) Choose different values of n_0 (such as 20, 30, etc) and compute and plot the amplitude spectrum |G(f)| of g[n] for -0.5 < f < 0.5.
- 4) Compare |G(f)| with |H(f)| and discuss the effect of n_0 on |G(f)|. Hint: you can plot |G(f)| on log scale vertically to see details of the small region in |G(f)|.
- 5) Also compute and plot the phase spectrum $\angle G(f)$ of g[n]. Is the phase spectrum always a linear function of f within -0.5 < f < 0.5? Why?
- 6) If the input applied to your designed filter is $x[n] = \cos(2\pi 0.1n) + 2\sin(2\pi 0.3n)$, what do you expect at the output of the filter? Can you verify it by computing y[n] = x[n] * g[n]?

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