

ASSIGNED: Apr. 9, 2015. **READ:** Sects. 11.1-11.4 if you haven't already.

DUE DATE: UNGRADED. **TOPICS:** Multirate filtering, DSP noise filtering.

NOTE: This problem set will not be collected or graded.

- [10] 1. *Using multirate filtering to alter the pitch of musical notes:*

You have a snippet of music at note E. You want to change note E to note A. Note A has frequency $2/3$ that of note E. Design a multirate system to do this.

- [30] 2. *Drill on multirate filtering and aliasing:*

A 400 Hertz sinusoid is input to the DSP systems below at $1000 \frac{\text{SAMPLE}}{\text{SECOND}}$.

The output of each one is one or more sinusoids. Compute their frequencies.

[10] a. 400 Hz \rightarrow $\boxed{\text{A/D}}$ \rightarrow $\boxed{\uparrow 2}$ \rightarrow $\boxed{\downarrow 3}$ \rightarrow $\boxed{\text{D/A}}$ \rightarrow ?

[10] b. 400 Hz \rightarrow $\boxed{\text{A/D}}$ \rightarrow $\boxed{\downarrow 3}$ \rightarrow $\boxed{\uparrow 2}$ \rightarrow $\boxed{\text{D/A}}$ \rightarrow ?

[10] c. 400 Hz \rightarrow $\boxed{\text{A/D}}$ \rightarrow $\boxed{\downarrow 4}$ \rightarrow $\boxed{\uparrow 2}$ \rightarrow $\boxed{\text{D/A}}$ \rightarrow ?

Try checking your answers using Matlab (see my lecture notes for Matlab commands).

Download `p0.mat` from the web site. `>>load p0.mat` to get sampled signals X_1, X_2, X_3 .

- [20] 3. X_1 is a periodic signal, with noise added, sampled at $1000 \frac{\text{SAMPLE}}{\text{SECOND}}$.

[5] a. Plot X_1 as a continuous-time signal (use `plot`, not `stem`).

[5] b. Compute its spectrum using `FX=fft(X1)/length(X1)`. Plot its magnitude for $0 \leq f \leq 20$ Hz. Note $f=(k-1)\frac{1000}{1000}=k-1$, so plot `abs(FX(1:20))` vs. `[0:19]`.

[5] c. *Threshold* its spectrum using `FX(abs(FX)<0.9)=0`. Plot it for $0 \leq f \leq 20$ Hz.

[5] d. Plot `length(X1)*real(ifft(FX))` for the *thresholded* FX . The noise is gone, even though we didn't know *anything* about the signal except that it was periodic. That told us its spectrum was a line spectrum, so we could eliminate the noise by setting everything that wasn't a large line to 0. Thresholding was an easy way.

- [20] 4. X_2 is a trumpet, with noise added, sampled at $44100 \frac{\text{SAMPLE}}{\text{SECOND}}$ (standard CD rate).

Repeat #3. Use a threshold of 0.0015, and plot the spectrum for $0 \leq f \leq 4000$ Hz.

Listen to the noisy and filtered trumpets. The noise is eliminated again!

- [20] 5. X_3 is 2 trumpets playing simultaneously notes A and B, sampled at $44100 \frac{\text{SAMPLE}}{\text{SECOND}}$.

[5] a. Plot `X3(1000:1199)` as a continuous-time signal (use `plot`, not `stem`).

[5] b. Plot the spectrum for $0 \leq f \leq 4000$ Hz. Note the pairs of harmonics.

[5] c. Knowing *only* that note B has a higher frequency than note A, eliminate the trumpet playing note A by setting its harmonics to 0.

[5] d. *Listen* to the result in the time domain. Only one trumpet is left!

“The value of an idea lies in the using of it”—Thomas Edison. DSP applications.