

1. You have access to a cheap analog low-pass filter which has a fairly flat unit gain response from  $\Omega=0$  to  $\Omega=60 \text{ KHz}$ , and a stopband attenuation of  $100 \text{ dB}$  for  $\Omega>80 \text{ KHz}$ . With this analog anti-aliasing filter you are asked to design a 16-bit A/D converter system with a baseband sampling rate at  $60 \text{ KHz}$ .
  - a) If you were to use the above analog filter for the design of an oversampling A/D converter system, what is the **minimum** sampling frequency that you should choose?
  - b) With the sampling rate that you picked in part (a), draw a block diagram of the A/D converter system from the analog input signal to the final sampled output sequence at  $60 \text{ KHz}$ . This block diagram needs to include the specification (passband bandwidth, stopband attenuation, gain, etc.) of any filter that would be needed in the system.
  - c) Is your answer in part (b) the most hardware-efficient solution? If not, use another sampling rate to design this A/D converter system with less amount of computation. Draw a block diagram to describe this system.

a): The minimum sampling rate is  $30\text{Khz} + 80\text{Khz} = 110 \text{ Khz}$

b): The block diagram consists of the above anti-aliasing filter, a sampler at  $110\text{Khz}$ , followed by an up-converter by a factor of 6, a digital filter, and a down-converter by a factor of 11. The digital filter specs are: passband,  $\pi/11$ ; stopband attenuation:  $100\text{dB}$ , gain: 6.

c):  $120\text{Khz}$ .