

## Solutions for Midterm #2 - EECS 145M Spring 2010

- 1.1** Frequency aliasing is caused by sampling at a frequency  $f_s$  that is less than twice the highest frequency  $f_{\max}$  in the waveform.

It may be avoided by using an analog low-pass filter to eliminate frequencies above  $f_s/2$ .

[4 points off for increasing the sampling frequency. When sampling an arbitrary waveform you don't know the maximum frequency.]

- 1.2** Spectral leakage is caused when frequency components are not sampled for a whole number of cycles, which results in a discontinuity between the last sample and the "next sample", which is also the first sample.

It is avoided by multiplying the sampled values by a windowing function that has zero value and zero slope at the ends of the sampling window.

[3 points off for a solution that samples for a longer time- this does not eliminate the discontinuity and long-range spectral leakage will still occur]

[3 points off if the only answer is to sample for an integer number of cycles- nothing is known about the frequency components present]

- 2.1** Filter gain  $>0.99$  for frequencies  $<78,400$  Hz

- 2.2** Filter gain  $<0.01$  for frequencies  $>177,800$  Hz

- 2.3**  $S = M \Delta t = M/f_s = 2^{16}/2^{18} \text{ Hz} = 0.25 \text{ s}$

- 2.4**  $H_0$  corresponds to 0 Hz (d.c.);  $H_1$  corresponds to  $1/S = 4$  Hz

- 2.5** The FFT produces coefficients  $H_n$ , where  $n = 0$  to  $M-1$ . Therefore, the coefficient with the highest index is  $H_{M-1}$  or  $H_{65,535}$ , which corresponds to 4 Hz.

[2 points off for  $H_M$  and 0 Hz] [3 points off for  $H_M$  and  $2^{18}$  Hz]

- 2.6** The FFT coefficient that corresponds to the highest frequency is  $H_{M/2}$  or  $H_{32,768}$ . The corresponding frequency is  $(M/2)/S = 131,072$  Hz

- 2.7** For a 4,000 Hz sinewave, the primary FFT coefficients are  $H_{1000}$  and  $H_{M-1000}$ . Additional neighboring coefficients  $H_{999}$ ,  $H_{1001}$ ,  $H_{M-999}$ , and  $H_{M-1001}$  are non-zero (actually half the value of the primary coefficients) due to the side lobes produced by the Hann window.

[2 points off for omitting side lobes] [2 points off for omitting  $H_{M-999}$ ,  $H_{M-1000}$ , and  $H_{M-1001}$ ]

- 2.8** For a 4,000 Hz symmetric square wave, a sequence of harmonics will appear at odd multiples of the 4,000 Hz fundamental. So  $H_{k1000}$  and  $H_{M-k1000}$  would be non-zero, and the Hann side lobes would be at  $H_{k1000-1}$ ,  $H_{k1000+1}$ ,  $H_{M-k1000-1}$ , and  $H_{M-k1000+1}$ .

[1 point off for omitting side lobes] [3 points off for omitting harmonics]

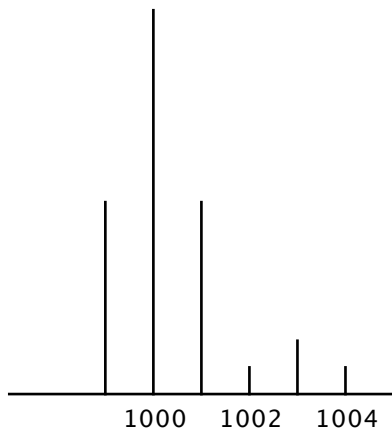
- 2.9** For a 4,002 Hz sinewave,  $H_{1000}$ ,  $H_{1001}$ ,  $H_{M-1000}$ , and  $H_{M-1001}$  would be non-zero and of equal magnitude, and the Hann side lobes would appear at  $H_{999}$ ,  $H_{1002}$ ,  $H_{M-999}$  and  $H_{M-1001}$ .

[1 point off for omitting side lobes] [2 points off for omitting  $H_{M-1000}$  and  $H_{M-1001}$ ]

[4 points off for stating that all coefficients are non-zero]

- 2.10** The primary 4,000 Hz sinewave would produce non-zero values at  $H_{999}$ ,  $H_{1000}$ , and  $H_{1001}$ . A second smaller sinewave of slightly higher frequency  $4,000 + 4m$  Hz would produce non-zero values at  $H_{1000+m-1}$ ,  $H_{1000+m}$ , and  $H_{1000+m+1}$  (there are also complex conjugate coefficients at  $H_{M-1000}$ , etc.). For the smaller sinewave to appear as a separate peak, there must be a valley between the coefficient  $H_{1001}$  and the coefficient at  $H_{1000+m}$ , which requires  $1000 + m > 1002$ , or  $m > 2$ . The smallest value of  $m$  we can have is 3, which corresponds to a frequency 12 Hz above 4,000 Hz.

[4 points off for 4 Hz] [3 points off for 8 Hz] [both 12 Hz and 16 Hz were accepted]



**2.11** A sinewave of frequency  $4M - 84,000 \text{ Hz} = 178,144 \text{ Hz}$  will produce non-zero coefficients at  $H_{20999}$ ,  $H_{21000}$ ,  $H_{21001}$ ,  $H_{M-20999}$ ,  $H_{M-21000}$ , and  $H_{M-21001}$ .

$$M = 2^{16} = 65,536. \quad M - 21,000 = 44,536.$$

A sinewave of frequency  $84,000 \text{ Hz}$  will produce non-zero coefficients at exactly the same frequency indexes. This is an example of how a higher frequency can alias to a lower frequency. However, the  $84,000 \text{ Hz}$  sinewave will be only slightly reduced by the anti-aliasing filter (gain  $>0.90$ , while the  $178,144 \text{ Hz}$  sinewave will be greatly reduced (gain  $\approx 0.01$ ). So the coefficients will be about 100 times smaller for the  $178,144 \text{ Hz}$  sinewave.

[3 points off for not stating the non-zero coefficients]

[1 point off for omitting  $H_{20999}$ ,  $H_{21001}$ ,  $H_{M-20999}$ , and  $H_{M-21001}$ ]

[3 points off for stating that the magnitudes are the same for sampling  $178,144$  and  $84,000 \text{ Hz}$ ]

**2.12** To reduce the answer to **2.10** by a factor of two (i.e. to  $6 \text{ Hz}$ ), sample for twice as long.

[2 points off for doubling the sampling frequency, which increases the number of Fourier coefficients but not the frequency spacing  $\Delta f = 1/S$ ]

**3.1** The Integral Fourier Transform will be zero except for integer multiples of the  $10 \text{ kHz}$  repeat frequency. The relative Fourier Amplitudes will depend on the waveform.

The lowest (1st) harmonic will appear at  $10 \text{ kHz}$

The highest (1000th) harmonic will appear at  $10 \text{ MHz}$

**3.2**  $M = 2000$  samples at  $f_s = 20 \text{ MHz}$  correspond to a sampling window  $S = M/f_s = 100 \mu\text{s}$ , which is one cycle of the waveform.

The Discrete Fourier Transform will have harmonics  $k = 1$  to  $1000$  at  $H_1$  to  $H_{1000}$  and their complex conjugates at  $H_{1001}$  to  $H_{1999}$

The first harmonic is  $H_1$  at  $10 \text{ kHz}$ , which is one cycle per  $100 \mu\text{s}$

The highest harmonic  $H_{1000}$  at  $10 \text{ MHz}$

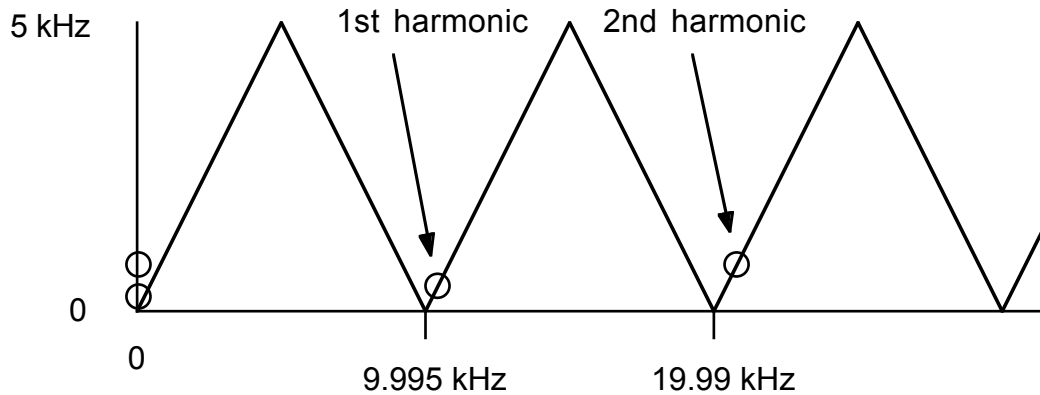
**3.3** Viewing this problem in the time domain the waveform has a period of  $1/10\text{kHz} = 100 \mu\text{s}$ . The sampling period of  $1/9.995\text{kHz} = 100.05 \mu\text{s}$ . Each sample will be progressively later by

0.05  $\mu$ s. The  $i$ th sample will be at time  $i \times 0.05\mu$ s. 2000 samples will sample one cycle of the waveform, and will be essentially the same as the values sampled in part 3.2

Viewing this problem in the frequency domain, the first harmonic at 10 kHz will be 0.005 kHz faster than 9.995 kHz (twice the sampling frequency) and alias to 0.005 kHz.

The  $k$ th harmonic at  $k \times 10$  kHz will be  $k \times 0.005$  kHz faster than  $k \times f_s$ .

The highest (1000th) harmonic at 10 MHz will 5 kHz faster than  $1000 \times f_s$  and will alias to 5 kHz.



Either approach was accepted for full credit.

Note: A sampling frequency of 9.995 kHz was chosen for this problem to avoid frequency overlap as described on page 397 of the textbook and on class handout slide 256.

**EECS145M Midterm #2 class statistics:**

Problem	max	average	rms
1	20	18.5	3.2
2	50	42.2	8.2
3	30	20.7	8.6
total	100	81.4	16.8

Grade distribution:

Range	number	approximate letter grade
30-39	1	F
40-49	0	F
50-59	0	D
60-69	2	C
70-79	4	B
80-89	3	A-, B+
90-99	5	A
100	2	A+