GPGN 404 Final Exam December 13, 2006

Name: _

Question:	1	2	3	4	5	6	Total
Points:	8	11	12	10	5	4	50
Score:							



Figure 1: A sequence x[n] and its amplitude spectrum. The sequence x[n] is not aliased and consists of 101 samples of a continuous signal $x_c(t)$.

(b) [2 points] What is the Nyquist frequency, in Hz?

(c) [2 points] What is the frequency (in Hz) of the periodic fluctuations in x[n].

(d) [2 points] Label the frequency axis below the amplitude spectrum.

- - (a) [2 points] Sketch the impulse response h[n] of this system.
 - (b) [2 points] Use a well-known formula to rewrite H(z) as the ratio of two polynomials in z.
 - (c) [2 points] For what two frequencies ω between 0 and π is the amplitude response $A(\omega)$ of this filter zero?
 - (d) [3 points] Assume that the sequence displayed in Figure 1 is $x[n] = cn + sin(\omega_0 n)$, for some constant c and frequency ω_0 . (You should already have determined the frequency ω_0 .) For this input x[n], and ignoring samples near the ends, what is the output y[n] of this moving average filter? Hint: the input x[n] to this LTI system is the sum of two sequences.

(e) [2 points] Write the main loop of a computer program that implements this filter.

- (a) [2 points] What is the frequency to be attenuated, in radians/sample?
- (b) [2 points] Sketch the locations of filter poles and zeros in the complex z-plane.

- (c) [2 points] What is the system function H(z) for your filter? (Include the region of convergence.)
- (d) [2 points] Modify your system function H(z) so that your filter does nothing at frequency zero (DC).
- (e) [2 points] Write a constant-coefficient difference equation relating filter output y[n] to input x[n].
- (f) [2 points] Write the main loop of a computer program that implements your filter.

- (a) [2 points] Why typically are only positive frequencies ω sampled in the array X[k]?
- (b) [2 points] Assuming that only positive frequencies ω are sampled, how many values are provided in the array X[k]?
- (c) [2 points] What is the frequency sampling interval $\Delta \omega$, in radians/sample?
- (d) [2 points] What is the index k_0 of the sample corresponding to the frequency nearest that of the periodic fluctuations in the sequence x[n]?
- (e) [2 points] Describe in words how you would use this index k_0 to implement a linear time-invariant filter that attenuates the periodic fluctuations in the sequence x[n]?

(a) [3 points] Implement the transformation $y_c(t) = x_c(\sqrt{t})$. That is, write an expression that exactly defines a new output sequence $y[n] = y_c(nT)$ in terms of the samples of the input sequence x[n].

(b) [2 points] In practice, we often sacrifice precision for efficiency in such transformations. How would you modify your computation of y[n] to make it faster, while still approximately correct?

(a) [2 points] Stable but not causal.

(b) [2 points] Causal but not stable.