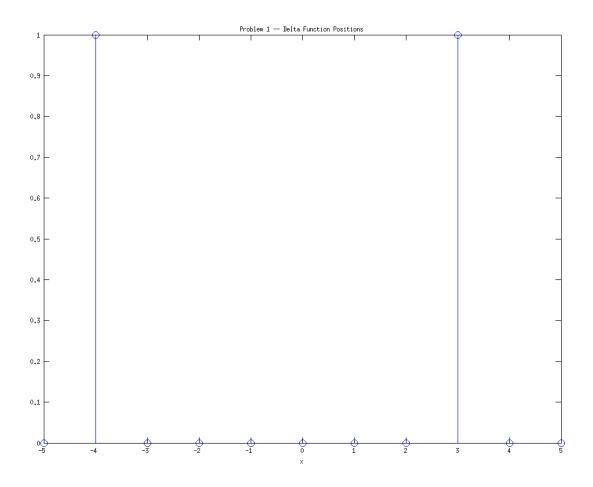
Advanced Digital Signal Processing Coursework Candidate Number: 137377

December 9, 2015

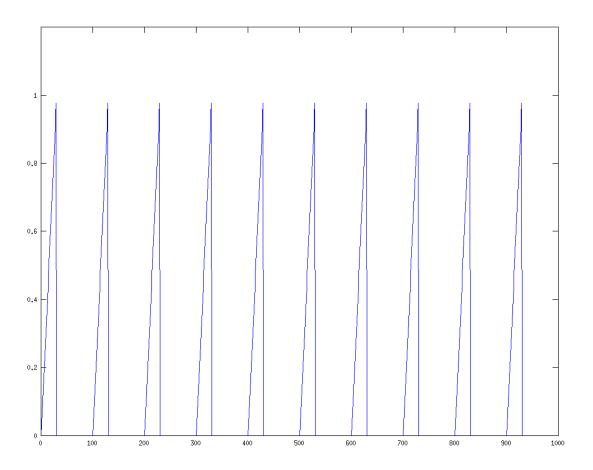
1 Problem 1a

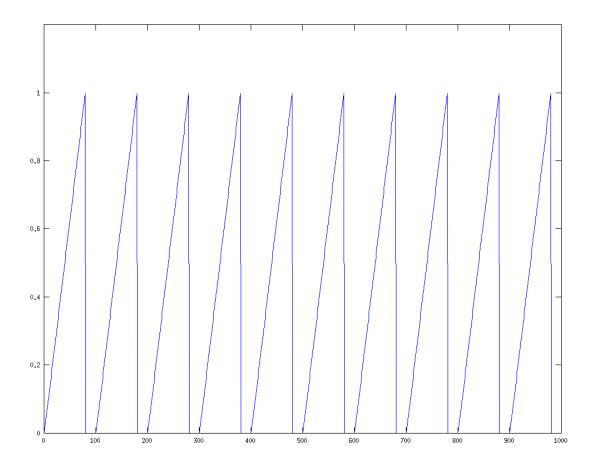


2 Problem 1b

```
1 % Problem # 1b (i & ii) Soln.:
       -----Sawtooth Function Plotting-
  %————Candidate No. 137377 —
5 %Generate a sawtooth waveform with a chosen duty cycle:
  clear all; close all;
  %Get the duty cycle percentage. Should be less than 100%.
  duty = mod(ceil(input('Enter desired Duty Cycle (%) : '))
      ,100);
<sup>11</sup> %Generate the sawtooth shape
  f = @(x) [x/duty.*(0 \le x & x \le duty)];
  x = linspace(0,100);
  y = f(x);
  %Number of times to repeat sawtooth in plot
  repetitions = 10;
  yy = y;
  for i=1:repetitions;
  yy = cat(2, yy, y);
  end
  % Plot
_{24} h = figure(1);
  plot(1:length(yy),yy)
26 ylim ([0 1.2]);
_{27} xlim([0 repetitions*100]);
```

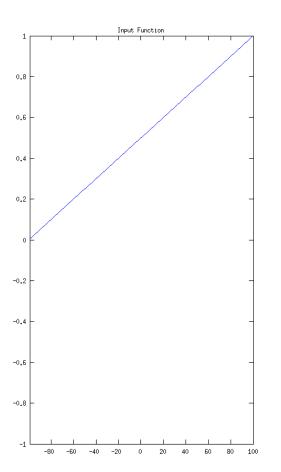
Showing output for 30% and 80% duty cycles respectively.

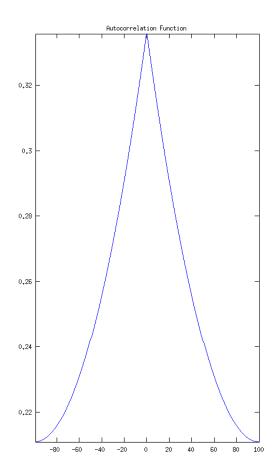




3 Problem 2(i)

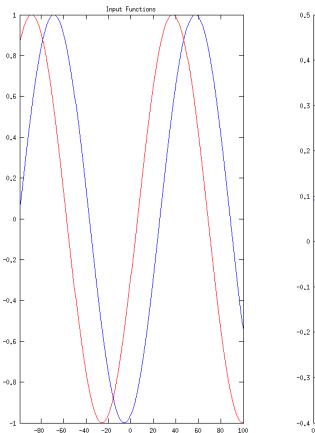
```
% Problem # 2(i) Soln.:
       -----Autocorrelation -
     ------Candidate No. 137377 ---
  %Auto-correlate a sequence with itself
  clear all; close all;
  N = 200;
  % Simple Line Input
  x = (1:200)/N;
11
  % Initialize array to hold the autocorrelation data
  psi = x*0;
  for i = 1:N
       x s hift = circ s hift (x, [0, (i + N/2)]);
17
       psi(i) = N^-1 * sum(x .* xshift);
18
  end
19
20
  h figure;
  subplot (1,2,1)
  plot(-floor(N/2)+1:floor(N/2),x);
  title ('Input Function');
  axis([-floor(N/2)+1 floor(N/2) -1 1]);
  subplot(1,2,2)
  plot(-floor(N/2)+1:floor(N/2), psi);
  title ('Autocorrelation Function');
  axis([-floor(N/2)+1 floor(N/2) min(psi) max(psi)])
```

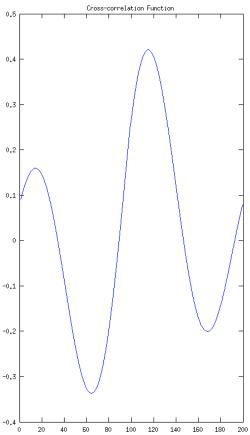




4 Problem 2(ii)

```
1 % Problem # 2(ii) Soln.:
        -----Cross Correlation -
      ------Candidate No. 137377 ----
  % Cross correlation of two sequences.
  clear all; close all;
  N = 200;
  % sin-shaped signal — slowly varying
  x(1:N) = \sin((1:N) / 20);
  y(1:N) = \sin(((1:N)+20) / 20);
  \% Initialize array to hold the correlation data
  psi = x*0;
  % Apply shift to cross-correlate
   for i = 1:N
       yshift = circshift(y, [0, (i + N/2)]);
       psi(i) = N^-1 * sum(x .* yshift);
19
  end
20
21
  h = figure
  subplot (1,2,1)
  plot(-floor(N/2)+1:floor(N/2),x,'b');
   title('Input Functions');
  axis([-floor(N/2)+1 floor(N/2) -1 1]);
  hold on;
  plot(-floor(N/2)+1:floor(N/2),y,'r');
  subplot(1,2,2)
  plot (1: size (psi, 2), psi);
  title ('Cross-correlation Function');
```

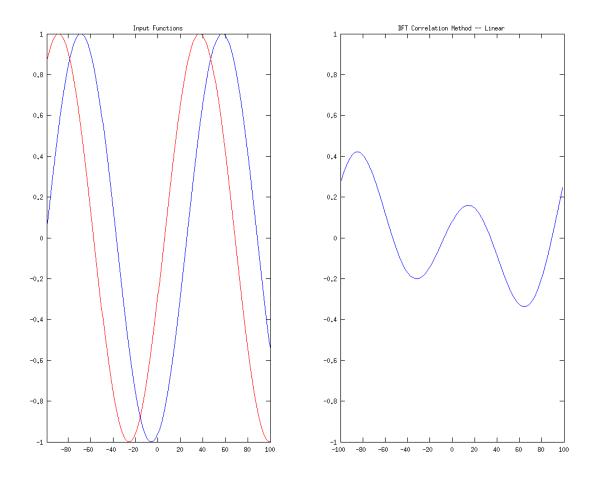


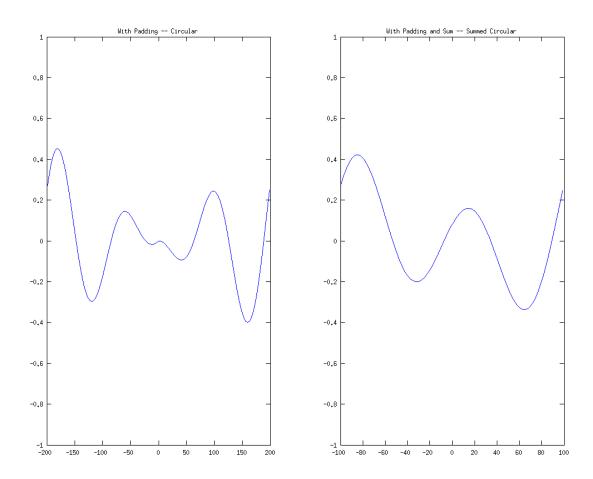


5 Problem 3

```
1 % Problem # 3 Soln.:
            ---Comparison of Correlation Methods-
  %
  %
       -----Candidate No. 137377 -
5 %
  % Graph 1 shows the input functions, Graph 2 shows the
      linear correlation, Graph 3 shows the circular
      correlation, Graph 4 shows the summed version of Graph
       3 to show that it is the same as Graph 2.
7 % Ultimately, the different methods produce the same
      output in Graphs 2 and
_{8} % 4. However, the zero-padding circular method produces
     an output that is
9 % periodic and must be summed to produce the final result
  clear all; close all;
11
  N = 200;
  % sin-shaped signal — slowly varying
  x(1:N) = \sin((1:N)/20);
  x_f = fft(fftshift(x));
  y(1:N) = sin(((1:N)+20) / 20);
  y_f = fft(fftshift(y));
  % Correlation without Padding
  % Do the multiplication in frequency space
  psi_f = x_f .* conj(y_f);
  psi_1 = (ifft(psi_f))/N;
  % Correlation with Zero Padding
  % Pad arrays to size N >= N_1 + N_2 - 1
  pad_size = size(x,2) + size(y,2) - 1;
  x_{padded} = zeros(1, pad_size); x_{padded}(size(x, 2) : end) = x
      (1:end);
  y_padded = zeros(1, pad_size); y_padded(1: size(y, 2)) = y
      (1:end);
  x_padded_f = fft(fftshift(x_padded));
30
  y_padded_f = fft(fftshift(y_padded));
  y_save = x_padded_f .* conj(y_padded_f);
  y_save = ifftshift(ifft(y_save));
```

```
psi_2 = y_save / N;
  psi_2\_sum = (y\_save(1:N) + y\_save(N:end)) / N;
38
  h = figure
  subplot (1,4,1)
  plot(-floor(N/2)+1:floor(N/2),x,'b');
   title('Input Functions');
  axis([-floor(N/2)+1 floor(N/2) -1 1]);
  hold on;
  plot(-floor(N/2)+1:floor(N/2),y, 'r');
  subplot (1,4,2)
  plot(-N/2:N/2-1, psi_1);
  title ('DFT Correlation Method -- Linear');
  axis([-N/2 \ N/2 \ -1 \ 1])
  subplot (1,4,3)
  plot(-N+1:N-1, psi_2); hold on;
  title ('With Padding — Circular')
  axis([-N N -1 1])
  subplot(1,4,4)
  plot(-N/2:N/2-1, psi_2\_sum);
  title ('With Padding and Sum -- Summed Circular')
  axis([-N/2 \ N/2 \ -1 \ 1])
```





6 Problem 4

```
1 % Problem # 4 Soln.:
            ---IIR Filter Design-
  %-
  %
  %————Candidate No. 137377 ——
  % Computes the frequency response through the bilinear
      transform method of
  % a high-pass IIR filter.
  % See report for the derivation of G.
  clear all; close all;
  % point_1_f = input('Enter desired -3dB cut-off frequency
     : ')
  % point_2 = input('Enter next attenuation point (dB): ')
  % point_2_f = input ('Enter desired frequency at the next
      attenuation point : ')
  % sample_rate = input('Enter sampling rate: ')
15
  % Inputs used for graph in report
  point_1_f = 1000;
  point_2 = 10;
  point_2 = 350;
  sample_rate = 5000;
  T = 1/sample_rate;
  w = 2*pi*(0:sample_rate/2);
  w_ac = (2/T) * tan(2*pi * point_1_f * T / 2);
  w_{att} = (2/T) * tan(2*pi * point_2_f * T / 2);
  filter_order = round(log10(10^(point_2/10) - 1) / (2*)
      log10(w_ac/w_att));
  disp(['Filter has order: ', num2str(filter_order)]);
  % Generate range of frequencies based on sample rate
  z = \exp(1 i *w*T);
  % Apply substitution for s for the type of filter, and
      the bilinear
  % transform
  s = ((2/T) .* ((z-1)./(z+1))).^filter_order;
  G = s \cdot / (s + (-1 \hat{ilter\_order}) * (w\_ac)^filter\_order);
  % High-pass filter:
```

```
<sup>39</sup>
<sup>40</sup> % Plot up until operating range of filter at z=-1
<sup>41</sup> h = figure;
<sup>42</sup> plot(w, abs(G).^2);
<sup>43</sup> xlabel('Frequency');
<sup>44</sup> ylabel('|G(f)|^2');
<sup>45</sup> title('Frequency Response of Filter')
```

The following were derived for use in the program:

Substitution for s in the high-pass filter case :
$$s = \frac{\omega_a}{s}$$
 (1)

$$G(s) = \frac{1}{1 + (-1)^n (\frac{\omega_{ac}^n}{s^n})}$$
 (2)

$$G(s) = \frac{s^n}{s^n + (-1)^n(\omega_{ac}^n)} \tag{3}$$

$$G(s) = \frac{1}{1 + (-1)^n \left(\frac{\omega_{ac}^n}{s^n}\right)}$$
(2)
$$G(s) = \frac{s^n}{s^n + (-1)^n (\omega_{ac}^n)}$$
(3)
Substitution for bilinear transform : $s = \frac{2}{T} \frac{z - 1}{z + 1}$ (4)

$$G(z) = \frac{\left(\frac{2}{T}\frac{z-1}{z+1}\right)^n}{\left(\left(\frac{2}{T}\frac{z-1}{z+1}\right)^n + (-1)^n(w_{ac})^n}$$
 (5)

(6)

Graph of the frequency response of the filter. With a -3dB cut-off frequency of 1 kHz, attenuation of 10 dB at 350 Hz, and a sampling frequency of 5 KHz.

