Part 1

Report Biomedical Signal Processing

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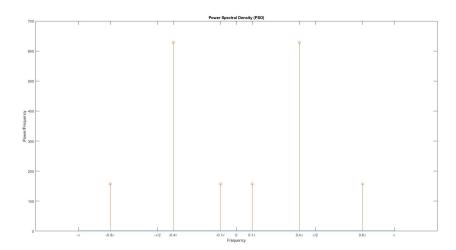
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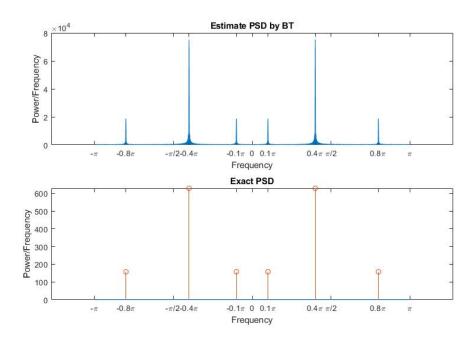
1 Question 1

1.1 part a

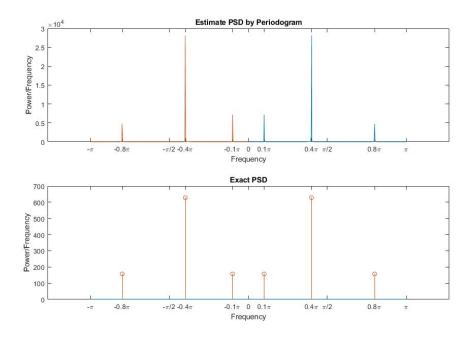
As we know for PSD of $Acos(\omega 0n + phi)$ is $A^2/2 * \pi(\delta(\omega - \omega 0) + \delta(\omega + \omega 0))$ and the PSD of white noise is σ^2 .



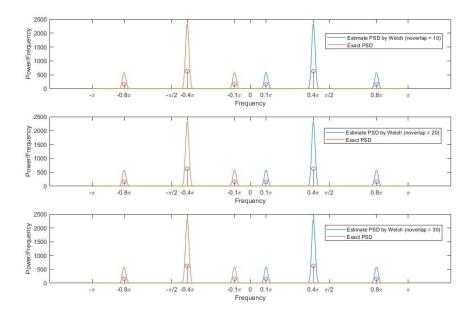
1.2 part b



1.3 part c

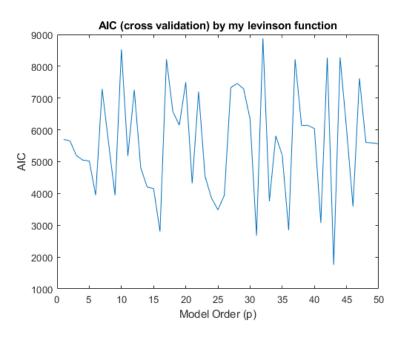


1.4 part d

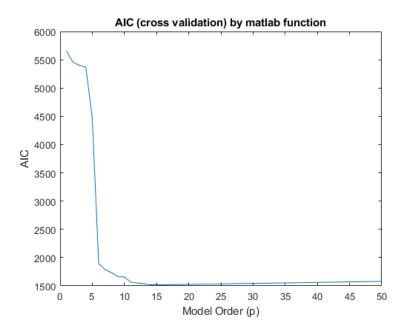


1.5 part e

- I have implemented my levonson function and matlab levin son and these are the results :
- By my function:



- By matlab function:



- optimal order:

```
Optimal AR coeffs:
   -0.1486
  -0.0970
   0.0584
   -0.1814
   -0.1533
    0.0999
   -0.0627
   -0.1252
    0.1621
    0.1041
   -0.0832
    0.1649
    0.1900
   -0.0650
    0.0728
   -0.8610
```

-coeffs for optimal order:

```
Optimal AR Order (p) by Levinson-Durbin:
18
Optimal AR Order (p) by AIC:
16
```

1.6 part f

1.7 part g

Matlab Code

```
1 % our information
  samples = 1000;
  alpha1 = 2*pi*rand;
_5 alpha2 = 2*pi*rand;
  alpha3 = 2*pi*rand;
n = 0: samples -1;
y = 10*\cos(0.1*pi*n + alpha1) + 20*\cos(0.4*pi*n + alpha1)
     alpha2) + 10*cos(0.8*pi*n + alpha3) + randn(1,
     samples);
10
11
 % part a
13
  f = linspace(-pi, pi, 1000);
  PSD = ones(size(f));
  add delta = @(psd, freq, amp) psd + amp * (f == freq);
 PSD = add delta(PSD, 0.1*pi, pi * 50);
  PSD = add delta(PSD, -0.1*pi, pi * 50);
 PSD = add delta(PSD, 0.4 * pi, pi * 200);
 PSD = add delta(PSD, -0.4 * pi, pi * 200);
  PSD = add delta(PSD, 0.8 * pi, pi * 50);
 PSD = add delta(PSD, -0.8 * pi, pi * 50);
  figure;
  plot(f, PSD);
  title ('Power Spectral Density (PSD)');
  xlabel('Frequency');
  ylabel('Power/Frequency');
28 hold on;
```

```
29 stem ([0.1*pi, -0.1*pi, 0.4*pi, -0.4*pi, 0.8*pi,
                    -0.8 * pi], \dots
                        [pi * 50, pi * 50, pi * 200, pi * 200, pi * 50, pi
30
                                        * 50]);
        x \operatorname{ticks} ([-pi -0.8*pi -pi/2 -0.4*pi -0.1*pi 0 0.1*pi
                    0.4*pi pi/2 0.8*pi pi]);
xticklabels (\{ '-\ pi', '-0.8\ pi', '-\ pi/2', '-0.4\ pi', '-10.4\ p
                    '-0.1\pi', '0', '0.1\pi', '0.4\pi', '\pi/2', '0.8\
                    pi', '\pi'});
33
34
35
       % part b
37
        estimated corr = zeros(1, samples/2);
       BT_PSD = zeros(1, 20000);
        f1 = linspace(-pi, pi, 20000);
41
        for i=1:samples/2
                         for j = 1:(samples-i)
                                         estimated corr(i) = x(j)*x(j+i) +
44
                                                    estimated corr(i);
                        end
45
                         estimated corr(i) = estimated corr(i) / samples;
46
        end
        for w=1:20000
                        temp = 0;
49
                         for j=1:(samples/2)
50
                                         temp = estimated corr(j) * exp(-1i*f1(w)*j) +
51
                                                    temp;
                        end
52
```

```
BT PSD(w) = estimated corr(1) + 2 * real(temp);
  end
55
  figure;
subplot (2,1,1);
  plot(fl, abs(BT PSD))
  title ('Estimate PSD by BT');
  xlabel('Frequency');
  ylabel('Power/Frequency');
 x \operatorname{ticks}([-pi -0.8*pi -pi/2 -0.4*pi -0.1*pi 0 0.1*pi
     0.4*pi pi/2 0.8*pi pi]);
  xticklabels (\{ '-\ pi', '-0.8\ pi', '-\ pi/2', '-0.4\ pi', 
     '-0.1\pi', '0', '0.1\pi', '0.4\pi', '\pi/2', '0.8\
     pi', '\pi'});
64
  subplot (2,1,2);
  plot(f, PSD);
67 hold on;
 title ('Exact PSD');
  xlabel('Frequency');
 ylabel('Power/Frequency');
  stem([0.1*pi, -0.1*pi, 0.4*pi, -0.4*pi, 0.8*pi,
     -0.8 * pi], \dots
      [pi * 50, pi * 50, pi * 200, pi * 200, pi * 50, pi
72
          * 50]);
r_3 xticks ([-pi -0.8*pi -pi/2 -0.4*pi -0.1*pi 0 0.1*pi
     0.4*pi pi/2 0.8*pi pi]);
xticklabels ({ '-\pi', '-0.8\pi', '-\pi/2', '-0.4\pi',
     '-0.1\pi', '0', '0.1\pi', '0.4\pi', '\pi/2', '0.8\pi
     pi', '\pi'});
75
```

```
76 %% part c
             [pxx, f2] = periodogram(x,[],[],2*pi);
           figure;
80
subplot (2,1,1);
            plot(f2, pxx); % Adjust frequency range to [-pi, pi]
           hold on;
             plot(-f2, pxx)
           title ('Estimate PSD by Periodogram');
           xlabel('Frequency');
          ylabel('Power/Frequency');
            x \text{ ticks} ([-pi -0.8*pi -pi/2 -0.4*pi -0.1*pi 0 0.1*pi
                              0.4*pi pi/2 0.8*pi pi]);
             xticklabels (\{ '-\pi', '-0.8\pi', '-\pi/2', '-0.4\pi', '-0.4\pi'
                                '-0.1\pi', '0', '0.1\pi', '0.4\pi', '\pi/2', '0.8\
                              pi', '\pi'});
90
             subplot (2,1,2);
            plot(f, PSD);
          hold on;
            stem([0.1*pi, -0.1*pi, 0.4*pi, -0.4*pi, 0.8*pi,
                               -0.8 * pi], \dots
                                      [pi * 50, pi * 50, pi * 200, pi * 200, pi * 50, pi
95
                                                             * 50]);
            title ('Exact PSD');
           xlabel('Frequency');
             ylabel('Power/Frequency');
            x \text{ ticks} ([-pi -0.8*pi -pi/2 -0.4*pi -0.1*pi 0 0.1*pi
                              0.4*pi pi/2 0.8*pi pi]);
             xticklabels (\{ '-\pi', '-0.8\pi', '-\pi/2', '-0.4\pi', '-0.4\pi'
```

```
'-0.1\pi', '0', '0.1\pi', '0.4\pi', '\pi/2', '0.8\
     pi', '\pi'});
101
102
103
104
  %% part d
106
107
108
  [pxx\_welch\_10, f3] = pwelch(x, 100, 10, samples, 2*pi)
     ;
  [pxx\_welch\_20, f4] = pwelch(x, 100, 20, samples, 2*pi)
  [pxx welch 30, f5] = pwelch(x, 100, 30, samples, 2*pi)
112
  figure;
113
  subplot(3,1,1);
  plot(f3, pxx_welch_10);
  hold on;
116
  plot(-f3, pxx welch 10)
117
  hold on;
  plot(f, PSD);
119
  hold on;
  stem([0.1*pi, -0.1*pi, 0.4*pi, -0.4*pi, 0.8*pi,
     -0.8 * pi], \dots
       [pi * 50, pi * 50, pi * 200, pi * 200, pi * 50, pi
122
           * 50]);
  xlabel('Frequency');
  ylabel('Power/Frequency');
```

```
legend ('Estimate PSD by Welch (noverlap = 10)', 'Exact
                                PSD')
                x \operatorname{ticks} ([-pi -0.8*pi -pi/2 -0.4*pi -0.1*pi 0 0.1*pi
                                 0.4*pi pi/2 0.8*pi pi]);
                xticklabels(\{ '-\pi', '-0.8\pi', '-\pi/2', '-0.4\pi', '-0.4\pi',
                                  '-0.1\pi', '0', '0.1\pi', '0.4\pi', '\pi/2', '0.8\
                                 pi', '\pi'});
128
              %%%%%%
129
130
                subplot (3,1,2);
131
               plot(f4 , pxx_welch_20);
132
                hold on;
                plot(-f4, pxx welch 20)
                hold on;
135
                plot(f, PSD);
               hold on;
137
                stem([0.1*pi, -0.1*pi, 0.4*pi, -0.4*pi, 0.8*pi,
138
                                  -0.8 * pi], \dots
                                        [pi * 50, pi * 50, pi * 200, pi * 200, pi * 50, pi
                                                               * 50]);
                xlabel('Frequency');
                ylabel('Power/Frequency');
              legend ('Estimate PSD by Welch (noverlap = 20)', 'Exact
                                PSD')
                x \operatorname{ticks} ([-pi -0.8*pi -pi/2 -0.4*pi -0.1*pi 0 0.1*pi
                                 0.4*pi pi/2 0.8*pi pi]);
                xticklabels(\{ '-\pi', '-0.8\pi', '-\pi/2', '-0.4\pi', '-0.4\pi',
                                  '-0.1\pi', '0', '0.1\pi', '0.4\pi', '\pi/2', '0.8\
                                 pi', '\pi'});
145
```

```
%%%%%%
147
148
  subplot(3,1,3);
149
  plot(f5, pxx welch 30);
150
  hold on;
151
  plot(-f5, pxx welch 30)
  hold on;
153
  plot(f, PSD);
154
  hold on;
  stem([0.1*pi, -0.1*pi, 0.4*pi, -0.4*pi, 0.8*pi,
      -0.8 * pi], \dots
       [pi * 50, pi * 50, pi * 200, pi * 200, pi * 50, pi
157
           * 50]);
  xlabel('Frequency');
158
  ylabel('Power/Frequency');
  legend('Estimate PSD by Welch (noverlap = 30)', 'Exact
     PSD')
  x \operatorname{ticks} ([-pi -0.8*pi -pi/2 -0.4*pi -0.1*pi 0 0.1*pi
      0.4*pi pi/2 0.8*pi pi]);
  xticklabels({ '-\pi', '-0.8\pi', '-\pi/2', '-0.4\pi',
      '-0.1\pi', '0', '0.1\pi', '0.4\pi', '\pi/2', '0.8\
      pi', '\pi'});
163
164
165
166
167
  % part e
168
170
```

```
R x = xcorr(x, 'biased');
172
  % implementing levinson by myself and AIC cross
173
      valiadation
_{174} E = [];
_{175} K=[];
  AIC = zeros(50,1);
  temp = 0;
  R_x = R_x(1000:1999);
178
   for i = 1:50
       if i == 1
180
            E(i) = R xx(i);
181
       elseif i == 2
182
            k(i) = -R_xx(i) / E(i-1);
183
            a(i,i) = k(i);
184
            E(i) = (1 - k(i)^2) * E(i-1);
185
       e1se
186
            for j = 1:i-1
187
                temp = temp + a(j, i-1) * R_x(i-j);
188
            end
189
            k(i) = -(R_xx(i) + temp) / E(i-1);
190
            a(i,i) = k(i);
191
            for j = 1:i-1
192
                 a(j,i) = a(j,i-1) + k(i) * a(i-j,i-1);
193
            end
194
            E(i) = (1 - k(i)^2) * E(i-1);
195
       end
196
       AIC(i) = samples * log(E(i)) + 2 * (i);
197
       temp = 0;
198
  end
  [\sim, p_opt_AIC] = min(AIC);
```

```
p_opt_LD = 1;
  disp('Optimal AR Order (p) by Levinson-Durbin:');
  disp(p opt LD);
203
  disp('Optimal AR Order (p) by AIC:');
  disp(p opt AIC);
205
  figure;
206
  plot (1:50, AIC);
   title ('AIC (cross validation)');
208
   xlabel('Model Order (p)');
209
  ylabel('AIC');
  disp('Optimal AR coeffs (p) by Levinson-Durbin:');
211
  disp(a(:,p_opt_LD));
212
  disp('Optimal AR coeffs (p) by AIC:');
  disp(a(:,p_opt_AIC));
  % implementing levinson by matlab function and AIC
      cross valiadation
216
  e = zeros(51,1);
217
  aic = zeros(50,1);
  [al, e(1)] = levinson(R xx, 0); % Order 0 model
  for p = 1:50
220
       [al, e(p+1)] = levinson(R xx, p);
221
       aic(p) = p*log(e(p+1))+2*p;
  end
223
  [\sim, p opt aic] = min(AIC);
  p opt LD m = 1;
  disp('Optimal AR Order (p) by Levinson-Durbin:');
  disp(a(:,p opt LD m));
  disp('Optimal AR Order (p) by AIC:');
  disp(a(:,p opt aic));
  figure;
```

```
plot (1:50, AIC);
   title('AIC (cross validation)');
   xlabel('Model Order (p)');
233
   ylabel('AIC');
235
236
  %% part f
238
239
  % Maximum MA order to consider
241
  max q = 20;
242
  % Preallocate arrays for criteria
   aic = zeros(max q, 1);
245
   bic = zeros(max_q, 1);
   fpe = zeros(max_q, 1);
247
248
  % Loop through different MA orders
   for q = 1: max q
       try
251
           % Estimate MA model of order q
252
            model = arima ('Constant', 0, 'MA', q, '
253
               Variance', 1);
            fit = estimate (model, x, 'Display', 'off', '
254
               EnforceInvertibility', true);
255
           % Get the log-likelihood, number of parameters
256
               , and variance of residuals
            logL = loglikelihood(fit);
257
            numParams = q + 1; % q MA parameters + 1
258
```

```
variance parameter
           variance = fit. Variance;
259
260
           % Calculate criteria
           aic(q) = -2 * logL + 2 * numParams;
262
           bic(q) = -2 * logL + log(N) * numParams;
263
           fpe(q) = variance * (1 + 2 * numParams / N) /
              (1 - 2 * numParams / N);
       catch ME
265
           % If estimation fails, set criteria to Inf
266
           aic(q) = Inf;
267
           bic(q) = Inf;
268
           fpe(q) = Inf;
269
           disp(['Order', num2str(q), 'estimation
               failed: ', ME. message]);
       end
  end
272
273
  % Find the order with minimum AIC, BIC, and FPE
  [\sim, \min aic order] = \min(aic);
  [\sim, \min bic order] = \min(bic);
  [\sim, \min fpe order] = \min(fpe);
  % Display results
  disp(['Optimal order by AIC: ', num2str(min_aic_order)
      1);
  disp(['Optimal order by BIC: ', num2str(min_bic_order)
   disp(['Optimal order by FPE: ', num2str(min fpe order)
      1);
283
```

```
% Plot AIC, BIC, and FPE
   figure;
285
   subplot (2,3,1);
286
   plot(1:max q, aic, 'LineWidth', 2);
   title ('AIC for Different MA Model Orders');
288
   xlabel('MA Order');
289
   ylabel('AIC');
   legend('AIC');
291
   grid on;
292
   subplot(2,3,2);
294
   plot(1:max q, bic, 'LineWidth', 2);
295
   title ('BIC for Different MA Model Orders');
   xlabel('MA Order');
297
   ylabel('BIC');
298
   legend('BIC');
   grid on;
301
   subplot(2,3,3);
   plot(1:max q, fpe, 'LineWidth', 2);
   title ('FPE for Different MA Model Orders');
304
   xlabel('MA Order');
   ylabel('FPE');
   legend('FPE');
307
   grid on;
308
309
310
311
312
313
314
```

```
315
316
    %% part g
317
319
320
322
323
324
325
326
327
328
329
330
331
332
    %% part h
```

2 Question 2

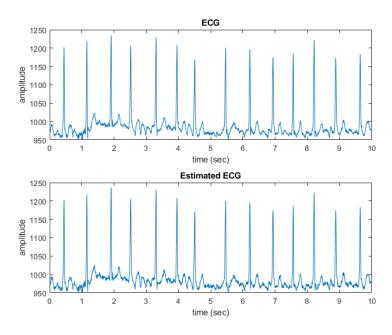
2.1 part a

```
1 %% a
2 ECG_dataset = load('test.mat');
3 fs = 360;
4 s = ECG_dataset.val;
5 t = 0:1/fs:(length(s)-1)/fs;
6 phi1 = 2*pi*rand;
7 phi2 = 2*pi*rand;
8 N1 = 2*cos(100*pi*t+phi1);
```

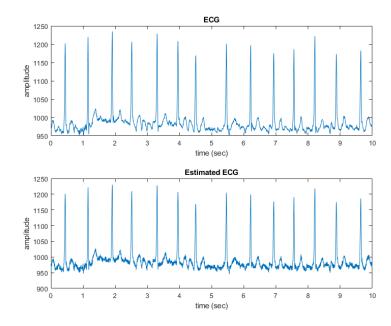
```
9 N2 = 2*cos(100*pi*t+phi2);
10 reference_signal = N1;
11 primary_signal = s + N2;
12 w = zeros(length(s),1);
13 y = zeros(length(s),1);
14 e = zeros(length(s),1);
15 mu = 0.000001;
16 p = y-w;
17 [e,y] = adaptivefilter(w, reference_signal, primary_signal, mu);
```

2.2 part b

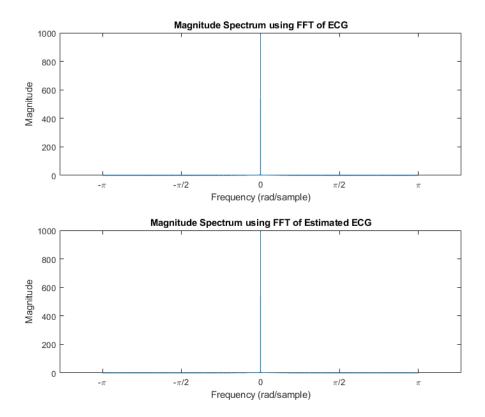
with Amplitude = 5 for noises :



with Amplitude = 20 for noises :



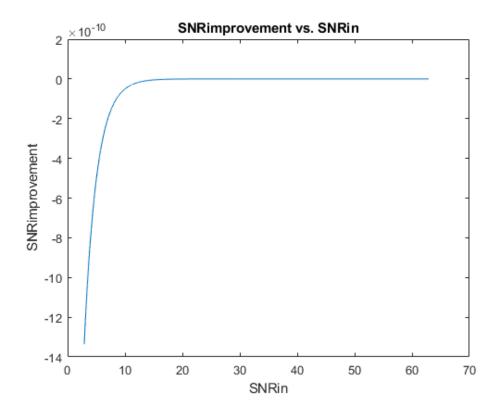
* As we can see still we have a little amount of effect from the noise, which added to the ECG signal, but we have estimated the ECG signal properly. Also we have to say that the effect of noise relate to the amplitude directly.



^{*} As we can see there is almost no difference in spectrum of the ECG and estimated ECG.

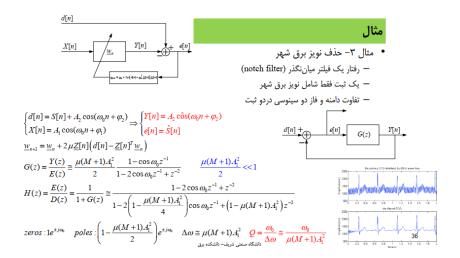
2.3 part c

- For different values of amplitude we have below figure, As we can see after some amplitudes the SNRimprovement will be constant even by increasing SNRin and converge.

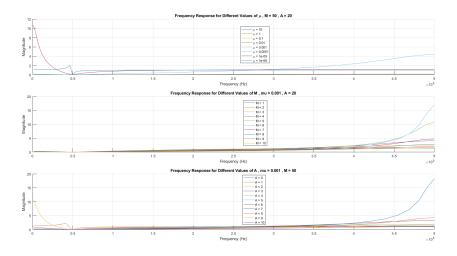


2.4 part d

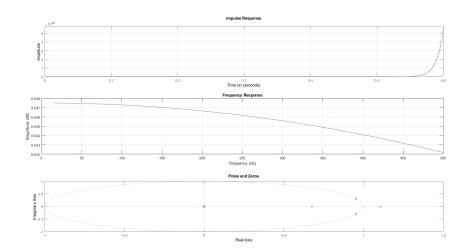
* As we had in slides:



- Now for different values of mu,A and M I've plotted frequency responses:

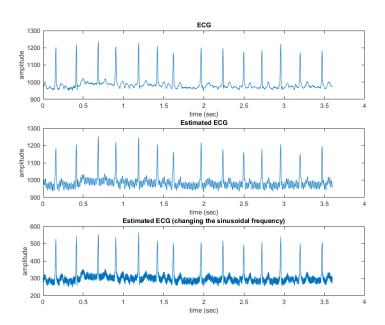


* Now for mu = 0.0001, A = 5, M = 100:



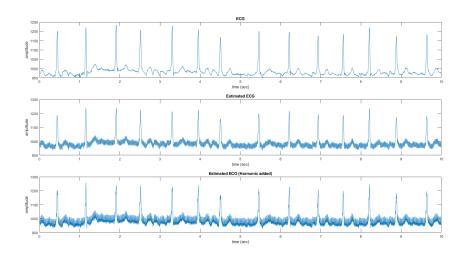
2.5 part e

- For analysis the effect of a little change in sinusoidal frequencies, I have added noise to the frequency and we will have :



^{*} As we can see the output of Adaptive filter would be noisier than previous estimated ECG.

2.6 part f



* As we can see by adding the first harmonic component Adaptive filter can remove less noise from primary signal but still estimation works almost properly.

Matlab Code

```
1 %% a
2 ECG_dataset = load('test.mat');
3 fs = 360;
4 s = ECG_dataset.val;
5 t = 0:1/fs:(length(s)-1)/fs;
6 phi1 = 2*pi*rand;
7 phi2 = 2*pi*rand;
8 N1 = 20*cos(100*pi*t+phi1);
9 N2 = 20*cos(100*pi*t+phi2);
10 reference_signal = N1;
11 primary_signal = s + N2;
12 w = zeros(length(s),1);
13 y = zeros(length(s),1);
14 e = zeros(length(s),1);
```

```
_{15} mu = 0.000001;
_{16} p = y-w;
 [e,y] = adaptivefilter(w, reference_signal,
     primary_signal ,mu);
18 % b
  figure;
20
  subplot(2,1,1)
  plot(t,s)
  title ('ECG');
  xlabel('time (sec)');
  ylabel('amplitude');
26
  subplot(2,1,2)
  plot(t,e)
  title ('Estimated ECG');
  xlabel('time (sec)');
  ylabel('amplitude');
 f = (-length(s)/2: length(s)/2-1)*(2*pi/length(s));
 figure;
  subplot (2,1,1)
  plot(f, abs(fftshift(fft(s)))/length(s));
  title ('Magnitude Spectrum using FFT of ECG');
  xlabel('Frequency (rad/sample)');
  ylabel('Magnitude');
  xticks([-pi -pi/2 0 pi/2 pi]);
  xticklabels({'-\pi', '-\pi/2', '0', '\pi/2', '\pi'})
42
 subplot(2,1,2)
```

```
plot(f, abs(fftshift(fft(e)))/length(e))
  title ('Magnitude Spectrum using FFT of Estimated ECG')
  xlabel('Frequency (rad/sample)');
  ylabel('Magnitude');
  x \operatorname{ticks}([-pi -pi/2 0])
                          pi/2
                                  pi]);
  xticklabels ({ '-\pi', '-\pi/2', '0', '\pi/2', '\pi'})
50
  %% c
52
  for i = 0:1000
53
      N1 = i*cos(100*pi*t+phi1);
54
      N2 = i*cos(100*pi*t+phi2);
55
      reference signal = N1;
56
       primary_signal = s + N2;
57
      w = zeros(length(s),1);
      y = zeros(length(s), 1);
59
      e = zeros(length(s), 1);
60
      mu = 0.000001;
61
      [e,y] = adaptive filter (w, reference signal,
62
          primary signal, mu);
      SNRin(i+1) = 10*log10(norm(s)^2/norm(N2)^2);
63
      SNRout(i+1) = 10*log10(norm(s)^2/(norm(e-s)^2));
      SNRimprovement(i+1) = SNRout(i+1) - SNRin(i+1);
65
  end
66
  figure;
  plot(SNRin, SNRimprovement)
  title ('SNRimprovement vs. SNRin');
  xlabel('SNRin');
```

```
ylabel('SNRimprovement');
73
74
  ‰ part d
76
77
  f = 50;
 z = tf('z', 1/fs);
  figure;
mu values = \begin{bmatrix} 10 & 1 & 0.1 & 0.01 & 0.001 & 0.0001 & 0.00001 \end{bmatrix}
     0.000001;
_{82} M = 50;
_{83} A = 20;
  for i = 1:length(mu_values)
      mu = mu values(i);
      H = (1 - 2*\cos(2*pi*f/fs)*z^{(-1)}+z^{(-2)}) / ...
86
           (1 - 2*(1 - (mu*(M+1)*A^2)/4)*cos(2*pi*f/fs)*z
               (-1) + (1 - mu*(M+1)*A^2)*z^(-2));
       subplot(3, 1, 1);
88
       hold on;
       [mag, phase, w] = bode(H);
       plot(w*fs/(2*pi), mag(:));
91
  end
  title ('Frequency Response for Different Values of \mu'
     );
  xlabel('Frequency (Hz)');
  ylabel('Magnitude');
  legend (arrayfun (@(x) ['\mu = 'num2str(x)], mu_values,
       'UniformOutput', false));
  grid on;
98
```

```
% Frequency response for different values of M
  mu = 0.001;
  M \text{ values} = 1:10;
101
  A = 20;
103
  for M = M values
104
       H = (1 - 2*\cos(2*pi*f/fs)*z^{(-1)}+z^{(-2)}) / ...
           (1 - 2*(1 - (mu*(M+1)*A^2)/4)*cos(2*pi*f/fs)*z
106
               (-1) + (1 - mu*(M+1)*A^2)*z^(-2));
       subplot(3, 1, 2);
107
       hold on;
108
       [mag, phase, w] = bode(H);
109
       plot (w*fs/(2*pi), mag(:));
110
  end
111
   title ('Frequency Response for Different Values of M');
112
  xlabel('Frequency (Hz)');
  ylabel('Magnitude');
  legend (array fun (@(x) ['M = 'num2str(x)], M_values, '
     UniformOutput', false));
  grid on;
  mu = 0.001;
117
  M = 50;
  A values = 0:10;
  for A = A values
120
       H = (1 - 2*\cos(2*pi*f/fs)*z^{(-1)}+z^{(-2)}) / \dots
121
           (1 - 2*(1 - (mu*(M+1)*A^2)/4)*cos(2*pi*f/fs)*z
122
               (-1) + (1 - mu*(M+1)*A^2)*z^(-2));
       subplot(3, 1, 3);
123
       hold on;
124
       [mag, phase, w] = bode(H);
       plot(w*fs/(2*pi), mag(:));
126
```

```
end
127
   title ('Frequency Response for Different Values of A');
  xlabel('Frequency (Hz)');
  ylabel('Magnitude');
  legend (array fun (@(x) ['A = 'num2str(x)], A_values, '
     UniformOutput', false));
  grid on;
  hold off;
133
134
  %%%%%
136
  mu = 0.0001;
137
  A = 5;
  M = 100;
  figure;
140
  subplot(3, 1, 1);
  impulse (H);
142
   title ('Impulse Response');
143
  xlabel('Time (s)');
  ylabel('Amplitude');
  grid on;
146
  subplot(3, 1, 2);
  [mag, phase, w] = bode(H, \{0, pi\});
  mag = squeeze(mag);
149
  w = squeeze(w);
  plot(w*fs/(2*pi), 20*log10(mag));
151
   title ('Frequency Response');
152
  xlabel('Frequency (Hz)');
  ylabel('Magnitude (dB)');
154
  grid on;
  subplot(3, 1, 3);
```

```
pzmap(H);
   title ('Poles and Zeros');
159
  % part e
161
162
164
165
166
  % part f
  phi1 = 2*pi*rand;
168
  phi2 = 2*pi*rand;
  phi3 = 2*pi*rand;
  N1 = 20*\cos(100*pi*t+phi1);
  N2 = 20*\cos(100*pi*t+phi2);
  first_harmonic = 20*cos(2*100*pi*t+phi3);
  reference_signal = N1+first_harmonic;
174
  primary signal = s + N2+first harmonic;
  w = zeros(length(s),1);
  y = zeros(length(s),1);
  e = zeros(length(s),1);
  mu = 0.000001;
  [eh, yh] = adaptive filter (w, reference signal,
      primary signal,mu);
  [ewh, ywh] = adaptive filter (w, reference signal -
     first_harmonic , primary_signal-first_harmonic ,mu);
  figure;
182
183
  subplot (3,1,1)
  plot(t,s)
```

```
title ('ECG');
   xlabel('time (sec)');
187
   ylabel('amplitude');
188
   subplot (3,1,2)
190
   plot(t,ewh)
191
   title ('Estimated ECG');
   xlabel('time (sec)');
193
   ylabel('amplitude');
194
   subplot (3,1,3)
196
   plot(t,eh)
197
   title ('Estimated ECG (Harmonic added)');
   xlabel('time (sec)');
199
   ylabel('amplitude');
200
201
202
203
  % part g
204
205
206
   phi1 = 2*pi*rand;
207
  N2 = 20*\cos(100*pi*t+phi2);
   primary_signal = s + N2;
209
  w = zeros(length(s), 1);
  mu = 0.000001;
   for i = 1:15
212
       [eal(i), yal(i)] = ALEfilter(w, primary_signal,
213
           primary signal, mu, i);
       SNRinal(i) = 10*log10(norm(s)^2/norm(N2)^2);
214
       SNRoutal(i) = 10*log10(norm(s)^2/(norm(e(i)-s)^2))
215
```

```
SNRimprovemental(i) = SNRoutal(i) - SNRinal(i);
216
   end
217
   figure;
   plot(SNRin, SNRimprovement)
219
   title ('SNRimprovement');
220
   xlabel('sample');
   ylabel('SNRimprovement');
222
  % Adaptive filter function:
223
   function [e,y] = adaptivefilter(w, reference_signal,
225
      primary signal, mu)
       for i=1:length(reference_signal)
           y = reference signal*w;
227
           e = primary signal - y;
228
           w = w + 2*mu*e*reference signal';
       end
230
   end
231
232
233
  % Adaptive Line Enhancer (ALE)
234
235
   function [e, y] = ALEfilter(w, reference signal,
      primary signal, mu, delay)
       % Initialize variables
237
       N = length (primary signal);
238
       y = zeros(1, N); % Filter output
239
       e = zeros(1, N); % Error signal
240
241
       % Apply delay to the primary signal
242
       delayed_primary_signal = [zeros(1, delay),
243
```

```
primary signal(1:end-delay)];
244
      % Adaptive filtering
245
       for i = 1:N
           % Ensure the reference signal is properly
247
              indexed to avoid overflow
           if i > delay
                ref signal segment = reference signal(i-
249
                   delay: i-1); % Delayed segment for
                   filter input
                primary segment = delayed primary signal(i
250
                   -delay: i-1); % Delayed primary segment
               y(i) = ref_signal_segment * w; % Filter
                   output
                e(i) = primary signal(i) - y(i); % Error
252
                   signal
               w = w + 2 * mu * e(i) * ref_signal_segment
253
                   '; % Update filter coefficients
           end
       end
255
  end
256
```

3 Question 3

3.1 part a

Article: "Model-based Prediction of Heart Rate Variability"

Summary: This study focuses on the use of parametric models for predicting heart rate variability (HRV) in clinical settings. The authors utilize autoregressive models to forecast HRV based on historical data, helping in the early detection of cardiac conditions. The approach allows for real-time monitoring and prediction of heart anomalies, providing valuable insights for preventive healthcare. By lever-

aging parametric modeling, the system adapts to individual patient data, enhancing the accuracy and reliability of predictions.

3.2 part b

Article: "Super-resolution spectral estimation in short-time

non-contact vital sign measurement" Summary: This study applies non-parametric spectral estimation methods to non-contact vital sign measurement using Doppler radar. Techniques such as short-time Fourier transform are employed to analyze the spectral content of radar signals reflected from the body. This allows for accurate extraction of vital signs like heart rate and respiratory rate in real-time, proving beneficial for remote patient monitoring and emergency medical applications.

3.3 part c

Article: "Parametric Spectral Estimation for Sleep Apnea Detection"

Summary: The research discusses the use of parametric spectral estimation techniques, specifically autoregressive (AR) modeling, to detect sleep apnea events from respiratory signals. By estimating the power spectral density of the respiratory signal, the method identifies characteristic patterns associated with apnea episodes. The parametric approach allows for high-resolution spectral analysis, making it possible to detect subtle changes in the signal that are indicative of apnea, thus aiding in the accurate diagnosis and monitoring of sleep disorders.

3.4 part d

Article: "Adaptive Filtering by Non-Invasive Vital Signals Monitoring and Diseases Diagnosis" Summary: The paper examines the use of adaptive filters, particularly LMS and RLS algorithms, in processing ECG and PPG signals. These filters dynamically adjust to varying noise conditions, effectively removing artifacts and enhancing signal quality. This facilitates accurate measurement of vital parameters such as heart rate and oxygen saturation, improving the reliability of non-invasive monitoring systems in clinical and home settings.