# Scilab Manual for Mobile Communication by Prof Hetal Shah Others Dharmsinh Desai University<sup>1</sup>

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May 26, 2017

<sup>&</sup>lt;sup>1</sup>Funded by a grant from the National Mission on Education through ICT, http://spoken-tutorial.org/NMEICT-Intro. This Scilab Manual and Scilab codes written in it can be downloaded from the "Migrated Labs" section at the website http://scilab.in



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# Digital Modulation Functions: ASK, FSK, PSK generation.

#### Scilab code Solution 1.1 1

```
1 // Amplitude Shift Keying, Frequency Shift Keying And
       Phase Shift keying waveform generation
2 clc;
3 clear;
4 xdel(winsid());
5 sym=10; //no. of symbols
6 g=[1 1 0 1 0 0 1 1 1 0 ]//binary data
7 f1=1;f2=2;//frequencies of carrier
8 t=0:2*%pi/99:2*%pi;//range of time
9 //ASK
10 cp=[]; bit=[]; mod_ask=[]; mod_fsk=[]; mod_psk=[]; cp1
      =[];cp2=[];
11 for n=1:length(g); //ASK modulation // Zeros and
      ones are inserted for proper plot of message
      signal
12
       if g(n) == 0;
           die=zeros(1,100);
13
14
       else g(n) == 1;
           die=ones(1,100);
```

```
16
       end
17
       c_ask=sin(f1*t);
18
       cp=[cp die];
19
       mod_ask=[mod_ask c_ask];
20
21
  ask=cp.*mod_ask;//ASK modulated signal
22
23 //FSK
24 for n=1:length(g);
25
       if g(n) == 0;
            die=ones(1,100);
26
27
            c_fsk=sin(f1*t);
28
       else g(n) == 1;
            die=ones(1,100);
29
30
            c_fsk=sin(f2*t);
31
       end
       cp1=[cp1 die];
32
33
       mod_fsk=[mod_fsk c_fsk];
34 end
35 fsk=cp1.*mod_fsk;//FSK molated signal
36
37 //PSK
38 for n=1:length(g);
       if g(n) == 0;
39
40
            die=ones(1,100);
41
            c_psk=sin(f1*t);
42
       else g(n) == 1;
            die=ones(1,100);
43
            c_psk = -sin(f1*t);
44
45
       end
       cp2=[cp2 die];
46
47
       mod_psk = [mod_psk c_psk];
48 end
49 psk=cp2.*mod_psk;//PSK modulated signal
50 subplot(4,1,1); plot(cp, 'LineWidth',1.5); // plot
      binary signal
51 xgrid;
52 title('Binary Signal');//title
```

```
53 mtlb_axis([0 100*length(g) -2.5 2.5]); //axis range
54 subplot (4,1,2); plot (ask, 'LineWidth',1.5); // plot of
     ASK modulated signal
55 xgrid;
56 title('ASK modulation');//title of plot
57 mtlb_axis([0 100*length(g) -2.5 2.5]);//axis range
58 subplot (4,1,3); plot (fsk, 'LineWidth',1.5); // plot of
     FSK modulated signal
59 xgrid;
60 title('FSK modulation');//title of plot
61 mtlb_axis([0 100*length(g) -2.5 2.5]);//axis range
62 subplot (4,1,4); plot (psk, 'LineWidth', 1.5); // plot of
     PSK modulated signal
63 xgrid;
64 title('PSK modulation');//title of plot
65 mtlb_axis([0 100*length(g) -2.5 2.5]);//range of
      axis
66 //Result: This experiment results plots of binary
      data, ASK modulation, FSK modulation and PSK
      modulation
```

Constellation diagram and Error Rate performance of different modulation techniques with AWGN channel.

Scilab code Solution 2.1 Sigal space diagram of different modulation

```
1 // Constellation diagram of BPSK and QPSK modulation
     and BPSK and QPSK modulation over AWGN channel
2 clc;
3 clear;
4 xdel(winsid());
5 sym=20; //No \cdot of symbols
6 data1=grand(1,sym,"uin",0,1);//Random symbol
     generation from 0 to 1 with uniform distribution
7 snr=10; // Signal to Noise Ratio
8 qpsk_mod = [];
9 bpsk_mod=2*data1-1; //BPSK Modulation
10 for j=1:2:length(data1)// Seperation of I & Q
     component for QPSK modulation
      i_phase=2*data1(j)-1;//BPSK modulation of I phase
11
          component
```

```
q_phase=2*data1(j+1)-1;//BPSK modulation of Q
12
         phase component
      temp=i_phase+%i*q_phase;//Combinibg I phase and Q
13
          phase component for QPSK modulation
14
      qpsk_mod=[qpsk_mod temp]; //QPSK modulated signal
15
16
     noise=1/sqrt(2)*(10^(-(snr/20)))*(rand(1,length(
17
        bpsk_mod), 'normal')+%i*(rand(1,length(bpsk_mod))
        , 'normal'))); // White gaussian noise generation
        for bpsk
18
     noise1=1/sqrt(2)*(10^(-(snr/20)))*(rand(1,length(
        qpsk_mod), 'normal')+%i*(rand(1,length(qpsk_mod)
        , 'normal'))); // White gaussian noise generation
        for apsk
     bpsk_awgn=bpsk_mod+noise; //BPSK Modulated signal
19
        passed over AWGN channel
20
    qpsk_awgn=qpsk_mod+noise1; //QPSK Modulated signal
       passed over AWGN channel
21
22
    figure // constellation diagram of ideal BPSK
       modulated signal and BPSK modulated signal with
       White Gaussian Noise
23 a = gca(); //to handle various object
24 a.data_bounds = [ -1 , -1;1 ,1];
25 a.x_location = "origin";
26 a.y_location = "origin";
27 plot2d ( real(bpsk_mod), imag(bpsk_mod), -2);
28 plot2d ( real(bpsk_awgn), imag(bpsk_awgn), -5);
29 xlabel( 'In phase'); //X-axis label
30 vlabel( 'Quadrature phase');//Y-axis label
31 title ('Constellation for BPSK with AWGN'); // title
      of plot
32 legend(['Ideal message point'; 'message point with
      noise ']);//legend
33 mtlb_axis([-2 2 -2 2]);//range of axis
34 figure // constellation diagram of ideal QPSK
     modulated signal and QPSK modulated signal with
```

```
White Gaussian Noise
35 a = gca(); //to handle various object
36 \text{ a.data\_bounds} = [ -1 , -1; 1 , 1];
37 a.x_location = "origin";
38 a.y_location = "origin";
39 plot2d ( real(qpsk_mod),imag(qpsk_mod),-2);
40 plot2d ( real(qpsk_awgn),imag(qpsk_awgn),-5);
41 xlabel ('In phase'); //X-axis label
42 ylabel ('Quadrature phase'); //Y-axis label
43 title( 'Constellation for QPSK with AWGN'); // title
      of plot
44 legend(['Ideal message point'; 'message point with
      noise ']);//legend
45 mtlb_axis([-2 2 -2 2]);//range of axis
46 //Result:Generates two plots: BPSK modulated signal
      with and without noise-figure -0
                         //QPSK modulated signal with
47
                            and without noise-figure-1
```

#### Scilab code Solution 2.2 BER of BPSK and QPSK over AWGN Channel

```
//Performance comparison of Simulated BER and
Theoritical BER of BPSK and QPSK modulation over
AWGN channel

clc;

clear;

xdel(winsid());

sym=10000; //No . of symbols

M=4;

qpsk_mod=[];i_phase=[];

data1=grand(1,sym,"uin",0,1); //Random Symbol
    generation from 0 to 1 with uniform distribution

for j=1:2:length(data1) // Seperation of I & Q
    component

i_phase=2*data1(j)-1; // BPSK modulation of I
```

```
phase component
      q_phase=2*data1(j+1)-1;//BPSK modulation of Q
11
         phase component
      temp=i_phase+%i*q_phase;//combining of I phase
12
         and Q phase component for QPSK modulation
      qpsk_mod=[qpsk_mod temp]; //QPSK modulated signal
13
14
   bpsk_mod=2*data1-1; //BPSK Modulated signal
15
16
17
      snr=1:10; // Signal to Noise Ratio
       for k=1:1:length(snr)
18
19
           H=1/sqrt(2)*(rand(1,length(qpsk_mod), 'normal
              ')+%i*(rand(1,length(qpsk_mod), 'normal'))
              );
           noise1=1/sqrt(2)*(10^(-(k/20)))*(rand(1,
20
              length(qpsk_mod), 'normal')+%i*(rand(1,
              length(qpsk_mod), 'normal')));//White
              Gaussian Noise generation for QPSK
           noise=1/sqrt(2)*(10^(-(k/20)))*(rand(1,
21
              length(bpsk_mod), 'normal')+%i*(rand(1,
              length(bpsk_mod), 'normal')));//White
              Gaussian Noise generation for QPSK
                   rec1_qpsk=qpsk_mod+noise1; //QPSK
22
                     modulated signal over AWGN channel
                   rec1_bpsk= bpsk_mod+noise; //BPSK
23
                     modulated signal over AWGN channel
24
25
                   rec_data_qpsk=[];rec_data_bpsk=[];
                  rec1_i=real(rec1_qpsk);//Seperation
26
                      of I phase and Q phase compnent
                      of received QPSK modulated signal
                   rec1_q=imag(rec1_qpsk);
27
28
           //
           for i=1:length(rec1_i)//QPSK Demodulation:
29
              BPSK demodulation of I phase and Q phase
              components
30
               if rec1_i(i)>=0
                     demod_out_i=1;
31
```

```
32
                 else rec1_i(i)<0</pre>
33
                      demod_out_i = 0;
34
                 end
35
                   if rec1_q(i) >= 0
36
                       demod_out_q=1;
37
                 else rec1_q(i)<0</pre>
38
                      demod_out_q=0;
                 end
39
                rec_data_qpsk=[rec_data_qpsk demod_out_i
40
                   demod_out_q];//QPSK Demodulated signal
41
                 end
             for i=1:1:length(data1)//BPSK Demodulation
42
43
                 if real(rec1_bpsk(i))>=0
                       demod_out_bpsk=1;
44
45
                 else real(rec1_bpsk(i))<0</pre>
                      demod_out_bpsk=0;
46
                 end
47
                 rec_data_bpsk = [rec_data_bpsk
48
                    demod_out_bpsk]; //BPSK Demodulated
                    signal
49
            end
50
51
            errA=0; errB=0;
52
        for i=1:sym
53
            if rec_data_qpsk(i) == data1(i)
54
                 errA = errA;
55
            else
                 errA = errA + 1;
56
57
            end
58
        end
            BER_qpsk(k)=errA/sym; // BER of QPSK
59
60
61
             for i=1:sym
            if rec_data_bpsk(i) == data1(i)
62
63
                 errB=errB;
64
            else
65
                 errB=errB+1;
66
            end
```

```
67
68
           BER_bpsk(k) = errB/sym; //BER of BPSK
69
       end
       theoryBer = 0.5*erfc(sqrt(10.^(snr/10))); //
70
          Theoritical BER of BPSK & QPSK
71
     end
72
          // end
73
74 snr=1:1:10;
75 plot2d(snr,BER_bpsk,5,logflag="nl");//plot simulated
      BER of BPSK over AWGN channel
76 plot2d(snr,BER_qpsk,2,logflag="nl");//plot simulated
      BER of QPSK over AWGN channel
77 plot2d(snr,theoryBer,3,logflag="nl");//Plot
      theoritical BER of QPSK and BPSK over AWGN
      channel
78 mtlb_axis([0 20 10^-5 0.5]); // axis
79 xgrid(10);
80 xtitle('Bit Error Rate plot for BPSK & QPSK
      Modulation', 'SNR', 'BER'); // title of plot
81
82
83 legend(['BER_sim_BPSK'; 'BER_sim_QPSK'; 'BER_Theory'])
      ;//legend
84 //This experiments results plot of bit error rate(
     BER) comparison of simulated BPSK over AWGN
      channel, simulated QPSK over AWGN channel and
      theoritical BER of BPSK and QPSK
85 // It will take few minutes to get plots as 100000
      bits are applied as an input to get better plots
```

# Effect of various channel on transmitted data using different modulation techniques.

Scilab code Solution 3.1 BER BPSK Rayleigh fading channel

```
1 //Error rate performance of BPSK modulated signal
      over only AWGN channel and AWGN and Rayleigh
      channel both
2 clc;
3 clear;
4 xdel(winsid());
5 sym=10000; //No \cdot of symbols
6 data1=grand(1,sym,"uin",0,1);//Randomly generated
      Symbolsfrom 0 to 1 with uniform distribution
8 bpsk_mod=2*data1-1; //BPSK Modulation
9 snr=1:20; //signal to Noise Ratio
       for k=1:1:length(snr)
10
11
12
           H1=1/sqrt(2)*(rand(1,length(bpsk_mod),'
              normal')+%i*(rand(1,length(bpsk_mod),'
              normal'))); // Rayleigh fading generation
```

```
13
14
           noise=1/sqrt(2)*(10^(-(k/20)))*(rand(1,
              length(bpsk_mod), 'normal')+%i*(rand(1,
              length(bpsk_mod), 'normal')));// White
              Gaussian Noise generation
15
16
                   rec1_bpsk=bpsk_mod+noise; //BPSK
                      modulated signal over AWGN channel
                   rec1_bpsk_ray1= H1.*bpsk_mod+noise;//
17
                     BPSK modulated signal over AWGN
                      channel and Rayleigh Fading
                      channel
18
                  rec1_bpsk_ray=conj(H1).*rec1_bpsk_ray1
                     ;//multiplication with conjugate of
                      rayleigh fading to nullify phase
                     because of Rayleigh Fading
                 // rec1_bpsk_ray=rec1_bpsk_ray1./(H1.*
19
                   conj(H1));
20
21
                   rec_data_bpsk=[];rec_ray_bpsk=[];
22
           for i=1:1:length(data1)//BPSK Demodulation
23
              of received signal over AWGN channel
                if real(rec1_bpsk(i))>=0
24
                     demod_out_bpsk=1;
25
                else real(rec1_bpsk(i))<0</pre>
26
27
                    demod_out_bpsk=0;
28
                end
                rec_data_bpsk=[rec_data_bpsk
29
                   demod_out_bpsk]; // Received signal
30
                if real(rec1_bpsk_ray(i))>=0 //BPSK
31
                   Demodulation of received signal over
                  AWGN channel and Rayleigh channel
                     demod_ray_bpsk=1;
32
                else real(rec1_bpsk_ray(i))<0</pre>
33
34
                    demod_ray_bpsk=0;
35
                end
```

```
36
                rec_ray_bpsk = [rec_ray_bpsk
                   demod_ray_bpsk]; ///Received signal
37
           end
38
39
           errB=0; errC=0;
40
       for i=1:sym
41
           if rec_data_bpsk(i) == data1(i) // Error rate
42
               calculation of received signal by
               considering only AWGN Channel
43
                errB=errB;
44
            else
45
                errB=errB+1;
46
            end
47
            BER_bpsk(k)=errB/sym; //BER at receiver by
48
               considering only AWGN Channel
49
             if rec_ray_bpsk(i) == data1(i) // Error rate
50
                calculation of received signal by
                considering AWGN Channel and Rayleigh
                channel
                errC=errC;
51
52
           else
53
                errC=errC+1;
54
           end
55
           BER_bpsk_ray(k) = errC/sym; //BER at receiver
56
              by considering AWGN Channel and rayleigh
              channel
57
       end end
58
59
          // end
60 snr=1:1:20;
61 plot2d(snr,BER_bpsk,5,logflag="nl");
62 plot2d(snr,BER_bpsk_ray,3,logflag="nl");
63 mtlb_axis([0 20 10^-5 0.5]);
64 xgrid(10);
```

```
65 xtitle( 'Bit Error Rate plot for BPSK modulated
      signal over AWGN channel and AWGN and Rayleigh
      channel both', 'SNR', 'BER');
66 legend(['BER_BPSK_AWGN'; 'BER_BPSK_AWGN & Rayleigh'])
   ;
67 //This experiment results plot of error rate
      performance of BPSK modulated signal over AWGN
      channe and AWGN and Rayleigh channel both.
68 //This experiment will take some time to display
      plot as higher no. of bits entered as an input to
      get better plots.
```

#### Scilab code Solution 3.2 BER QPSK Rayleigh channel

```
1 //Error rate performance of QPSK modulated signal
     over only AWGN channel and AWGN and Rayleigh
     channel both
2 clc;
3 clear;
4 xdel(winsid());
5 sym=10000; //No.of.symbols
6 M=4;
8 data1=grand(1,sym,"uin",0,1);//Random Symbol
     generation from 0 to 1 with uniform distribution
9 for j=1:2:length(data1)// Seperation of I & Q
     component
10
     i_phase=2*data1(j)-1;// BPSK modulation of I
        phase component
     q_phase=2*data1(j+1)-1;//BPSK modulation of Q
11
        phase component
     temp=i_phase+%i*q_phase;//combining of I phase
12
        and Q phase component for QPSK modulation
13
     qpsk_mod=[qpsk_mod temp]; //QPSK modulated signal
14
   end
```

```
15
16
      snr=1:5:41; // Signal to Noise Ratio
       for k=1:length(snr)
17
           H=1/sqrt(2)*(rand(1,length(qpsk_mod), 'normal
18
              ')+%i*(rand(1,length(qpsk_mod),'normal'))
              );//Rayleigh fading generation
19
           noise1=1/sqrt(2)*(10^(-(k/20)))*(rand(1,
20
              length(qpsk_mod), 'normal')+%i*(rand(1,
              length(qpsk_mod), 'normal')));//White
              Gaussian Noise generation for QPSK
21
22
                   rec1_qpsk=qpsk_mod+noise1; //QPSK
                      modulated signal over AWGN channel
                    rec1_qpsk_ray1= H.*qpsk_mod+noise1;
23
                      //BPSK modulated signal over AWGN
                        channel and Rayleigh Fading
                      channel
                   rec1_qpsk_ray=conj(H).*rec1_qpsk_ray1
24
                      ;//multiplication with conjugate
                      of rayleigh fading to nullify
                      phase because of Rayleigh Fading
25
                   rec_data_qpsk=[]; rec_data_qpsk_ray
26
                     =[];
27
28
                   rec1_i=real(rec1_qpsk);//Seperation
                      of I phase and Q phase comppnent
                      of received QPSK modulated signal
29
                   rec1_q=imag(rec1_qpsk);
30
31
                    rec1_i_ray=real(rec1_qpsk_ray);//
                      Separation of I phase and Q phase
                       component of received QPSK
                      modulated signal
32
                   rec1_q_ray=imag(rec1_qpsk_ray);
33
           for i=1:length(rec1_i)//QPSK Demodulation:
34
```

```
BPSK demodulation of I phase and Q phase
               components
                 if rec1_i(i)>=0
35
36
                       demod_out_i=1;
37
                 else rec1_i(i)<0</pre>
38
                      demod_out_i=0;
39
                 end
40
                   if rec1_q(i) >= 0
                       demod_out_q=1;
41
42
                 else rec1_q(i)<0</pre>
                      demod_out_q=0;
43
44
                  end
45
                     if rec1_i_ray(i) >= 0
                       demod_out_i_ray=1;
46
47
                 else rec1_i(i)<0</pre>
48
                     demod_out_i_ray=0;
49
                 end
                   if rec1_q_ray(i) >= 0
50
                       demod_out_q_ray=1;
51
                 else rec1_q_ray(i)<0</pre>
52
                     demod_out_q_ray=0;
53
54
                 end
                rec_data_qpsk=[rec_data_qpsk demod_out_i
55
                   demod_out_q]; //QPSK Demodulated signal
                rec_data_qpsk_ray=[rec_data_qpsk_ray
56
                   demod_out_i_ray demod_out_q_ray]; //
                   QPSK Demodulated signal
57
                 end
58
            errA=0; errB=0;
59
        for i=1:sym
60
61
            if rec_data_qpsk(i) == data1(i)
62
                 errA = errA;
63
            else
64
                 errA = errA + 1;
65
            end
66
        end
            BER_qpsk(k)=errA/sym; // BER of QPSK
67
```

```
68
             for i=1:sym
69
70
            if rec_data_qpsk_ray(i) == data1(i)
                errB=errB;
71
72
            else
73
                errB=errB+1;
74
            end
75
            BER_qpsk_ray(k)=errB/sym;//BER of BPSK
76
77
       end
       //\text{theoryBer} = 0.5 * \text{erfc} ( \text{sqrt} (10.^{\circ} ( \text{snr} / 10) ) ); //
78
          Theoritical BER of BPSK & QPSK
79
     end
80
81
           // end
82 snr=1:5:41;
83 plot2d(snr,BER_qpsk,5,logflag="nl");//plot simulated
       BER of BPSK over AWGN channel
84 plot2d(snr,BER_qpsk_ray,2,logflag="nl");//plot
      simulated BER of QPSK over AWGN channel
85 //plot2d (snr, theoryBer, 3, logflag="nl");//Plot
      theoritical BER of QPSK and BPSK over AWGN
      channel
86 mtlb_axis([0 40 10^-5 0.5]); // axis
87 xgrid(10);
88 xtitle('Bit Error Rate plot for QPSK over AWGN
      channel & AWGN and Rayleigh channel both', 'SNR',
       'BER') ;//title of plot
89
90 legend(['BER_QPSK_AWGN'; 'BER_QPSK_AWGN & Rayleigh'])
      ;//legend
91 //This experiments results plot of bit error rate (
      BER) comparison of simulated QPSK over AWGN
      channel, simulated QPSK over AWGN channel and
      Rayleigh fading channel.
92 // It will take few minutes to get plots as 10000
      bits are applied as an input to get better plots
```

#### Scilab code Solution 3.3 1

```
1 //Error rate performance of BPSK modulated signal
     over only AWGN channel and AWGN and Rayleigh
     channel both
2 clc;
3 clear;
4 xdel(winsid());
5 sym=10000; //No.of.symbols
6 data1=grand(1,sym,"uin",0,1);//Randomly generated
     Symbolsfrom 0 to 1 with uniform distribution
7
8 bpsk_mod=2*data1-1; //BPSK Modulation
9 snr=1:20; //signal to Noise Ratio
       for k=1:1:length(snr)
10
11
           H1=1/sqrt(2)*(rand(1,length(bpsk_mod),'
12
              normal')+%i*(rand(1,length(bpsk_mod),'
              normal')));//Rayleigh fading generation
13
14
           noise=1/sqrt(2)*(10^(-(k/20)))*(rand(1,
              length(bpsk_mod), 'normal')+%i*(rand(1,
              length(bpsk_mod), 'normal')));// White
              Gaussian Noise generation
15
16
                  rec1_bpsk=bpsk_mod+noise; //BPSK
                     modulated signal over AWGN channel
                  rec1_bpsk_ray1= H1.*bpsk_mod+noise;//
17
                     BPSK modulated signal over AWGN
                     channel and Rayleigh Fading
                     channel
                 rec1_bpsk_ray=conj(H1).*rec1_bpsk_ray1
18
                    ;//multiplication with conjugate of
                     rayleigh fading to nullify phase
```

```
because of Rayleigh Fading
                     rec1_bpsk_ray = rec1_bpsk_ray1./(H1.*
19
                    conj (H1));
20
21
                   rec_data_bpsk=[]; rec_ray_bpsk=[];
22
            for i=1:1:length(data1)//BPSK Demodulation
23
               of received signal over AWGN channel
24
                if real(rec1_bpsk(i))>=0
                      demod_out_bpsk=1;
25
                else real(rec1_bpsk(i))<0</pre>
26
                    demod_out_bpsk=0;
27
28
                end
29
                rec_data_bpsk=[rec_data_bpsk
                   demod_out_bpsk]; // Received signal
30
                if real(rec1_bpsk_ray(i))>=0 //BPSK
31
                   Demodulation of received signal over
                   AWGN channel and Rayleigh channel
                      demod_ray_bpsk=1;
32
33
                else real(rec1_bpsk_ray(i))<0</pre>
                    demod_ray_bpsk=0;
34
35
                end
                rec_ray_bpsk = [rec_ray_bpsk
36
                   demod_ray_bpsk]; ///Received signal
37
            end
38
            errB=0; errC=0;
39
       for i=1:sym
40
41
42
            if rec_data_bpsk(i) == data1(i) // Error rate
               calculation of received signal by
               considering only AWGN Channel
                errB=errB;
43
44
            else
45
                errB=errB+1;
            end
46
47
```

```
BER_bpsk(k) = errB/sym; //BER at receiver by
48
              considering only AWGN Channel
49
50
            if rec_ray_bpsk(i) == data1(i) // Error rate
               calculation of received signal by
               considering AWGN Channel and Rayleigh
               channel
               errC=errC;
51
52
           else
53
               errC=errC+1;
54
           end
55
56
           BER_bpsk_ray(k) = errC/sym; //BER at receiver
              by considering AWGN Channel and rayleigh
              channel
57
       end
            end
58
59
          // end
60 snr=1:1:20;
61 plot2d(snr,BER_bpsk,5,logflag="nl");
62 plot2d(snr, BER_bpsk_ray, 3, logflag="nl");
63 mtlb_axis([0 20 10^-5 0.5]);
64 xgrid(10);
65 xtitle ('Bit Error Rate plot for BPSK modulated
      signal over AWGN channel and AWGN and Rayleigh
      channel both', 'SNR', 'BER');
66 legend(['BER_BPSK_AWGN'; 'BER_BPSK_AWGN & Rayleigh'])
67 //This experiment results plot of error rate
      performance of BPSK modulated signal over AWGN
      channe and AWGN and Rayleigh channel both.
68 //This experiment will take some time to display
      plot as higher no. of bits entered as an input to
       get better plots.
```

Trunking Theory for Probability of blocking(Erlang B) and probability of delay(Erlang C).

Scilab code Solution 4.1 Traffic calculation in Erlang B and Erlang C system

```
1 //Exp-4 Calculates maximum traffic intensity and
     maximum no. of users accomodated in Erlang B and
     Erlang C system for given no of channels
2 clc;
3 clear;
4 xdel(winsid());
6
    function [p1]=erlangB(A1,c1)// calculate blocking
       probability for Erlang B system
7
         pr2=0;
8
           pr1=A1^c1/factorial(c1);
           for k=1:c1
9
10
           pr2=pr2+(A1^k/factorial(k));
11
           end
```

```
// A1=A1+1;
12
           p1=pr1/pr2;
13
     endfunction
14
15
16
     function [p2] = erlangC(A2,c2) // calculate
        probability of blocked call delayed in Erlang C
         system
         temp_1=0;
17
     for k=0:c2-1
18
       temp_1=temp_1+A2^k/factorial(k);
19
20 end
denominator=A^c2+(factorial(c2)*(1-(A2/c))*temp_1);
22 p2=A2^c2/denominator;
23 endfunction
24
25 pr_blocking=input('enter probability of blocking');
      //enter probability of blocking for perticular
     system
26 pr_delay=input('enter probability of block call
      delay'); //enter probability of blocked call
      delayed for particular system
27 y=input('enter call rate'); // Average no .of calls
      per minute
28 H=input('enter the average call duration'); //
      Average call duration in minute
29 c=input("enter no. of channels"); // Enter no. of
      channels
30 disp("no.of channel=");
31 disp(c);
32 Au=y*H; // Traffic intensity per user
33
34
   p=0;
        for A=1:1:100
35
            while (p<pr_blocking) // Find maximum traffic
36
               intensity for entered blocking
               probability pr_blocking
           [p]=erlangB(A,c)//calling function erlangB
37
           A = A + 1;
38
```

```
39
           end
         disp(pr_blocking, 'for blocking probability of'
40
            );//display blocking probability
         disp(A-1, 'Maximum traffic intensity is');//
41
            display max. traffic intensity
          u=(A-1)/Au;//no. of users calculation
42
         disp(u, "no .of users are accommodated");//
43
            display maximum no. of users accomodated in
            Erlang B system
         break;
44
     end//
45
     ; 0=q
46
47
         for A = 1:1:100
         while(p<pr_delay)//Find maximum traffic</pre>
48
            intensity for entered blocking probability
            pr_blocking
            [p]=erlangC(A,c)//calling funtion to
49
              calculate erlang C probability
           A = A + 1;
50
51
       end
52
         disp(pr_delay, 'for block call delay
            probability of'); // display blocking
            probability
         disp(A-1, 'Maximum traffic intensity is');//
53
            display max. traffic intensity
         u=(A-1)/Au;
54
         disp(u, "no. of users are accommodated");//
55
            display maximum no. of users accomodated in
            Erlang C system
         break;
56
57
     end
    //Enter blocking probability pr_blocking = 0.01
58
    //Enter probabolity of block call delay pr_delay
59
       =0.1
    //Enter call rate= 3/60
60
    //enter call duration= 2(in minute)
61
62
    //Enter no of channels 50
63
```

```
//Output:
64
   //no.of channel = 50.
65
66
   // for blocking probability of 0.01
67
    // Maximum traffic intensity is 38.
68
    // no .of users are accomodated 380.
69
70
    // for block call delay probability of 0.1
71
    // Maximum traffic intensity is 41.
72
    //no.of users are accommodated
73
```

### Walsh Code generation

Scilab code Solution 5.1 Walsh code generation and spreading and despreading using Walsh code

```
1 // Walsh Code generation
2 //Spreading and despreading of information for three
       users using Walsh code
3 clc;
4 clear;
5 xdel(winsid());
6 a=input('enter the number order of 2:');//input
      required length of Walsh Code which is always
      order of 2
7 c1=[1 -1 -1];//information of user 1
8 c2=[-1 \ 1 \ -1];//information of user 2
9 c3=[1 -1 1];//information of user 3
10 W=[0 0;0 1]; // Basic Walsh code Matrix
11 m=2;
12 %n=2^m;
13 \text{ for m} = 2:1:a
14 for i = 1:1:a//genration of walsh code matrix of
      entered length
       if i==2^m
15
       Winv=bitcmp(W,1);
16
```

```
W = [W W; W Winv];
17
18
           end
19
20 end
21 end
22 temp=0;
23 \text{ W1} = [];
24 disp(W)
25 for i=1:1:length(W(1,:))//0 replaced by -1 in walsh
      code matrix
       for j=1:1:length(W(1,:))
26
           if W(i,j) == 0 then
27
28
               W(i,j) = W(i,j) - 1;
            else W(i,j)=W(i,j)+0;
29
30
31
        end
32
33 end
34
35 end
36 // disp (W)
37 //spreading using Walsh code
38 tans_c1 = [c1(1,1).*W(1,:) c1(1,2).*W(1,:) c1(1,3).*W
      (1,:)];//spreading of user 1 information using
      first row of Walsh Matrix
39 tans_c2=[c2(1,1).*W(2,:) c2(1,2).*W(2,:) c2(1,3).*W
      (2,:)];//spreading of user 2 information using
      second row of Walsh Matrix
40 tans_c3=[c3(1,1).*W(3,:) c3(1,2).*W(3,:) c3(1,3).*W
      (3,:)];//spreading of user 3 information using
      third row of Walsh Matrix
41 aa1=tans_c1(1,1:a)+tans_c2(1,1:a)+tans_c3(1,1:a);
42 \quad aa2=tans_c1(1,(a+1):(2*a))+tans_c2(1,(a+1):(2*a))+
      tans_c3(1,(a+1):(2*a));
  aa3=tans_c1(1,((2*a))+1:(3*a))+tans_c2(1,((2*a))
      +1:(3*a))+tans_c3(1,((2*a))+1:(3*a));
44 tans_sig=[aa1 aa2 aa3];//transmission of spreaded
      signal
```

```
45 det_code1=input('enter detection code'); //Enter any
      integer no. ranging up to no. of rows of walsh
      matrix
46
47
       select det_code1//select case to get information
           of entered user
       case 1
48
            det_code=W(1,:);
49
50
       case 2
            det_code=W(2,:);
51
52
        case 3
53
            det_code=W(3,:);
54
       else
           det_code=W(4,:);
55
           disp('invalid detection code');//display
56
              message for input of invalid detection
             code
57
            end
58
59
60 rec_sig =[det_code(1,:).*aa1 det_code(1,:).*aa2
      det_code(1,:).*aa3];//received signal multiplied
      with detection code
61 \det_{\text{sig}} = [\text{rec\_sig}(1,1) + \text{rec\_sig}(1,2) + \text{rec\_sig}(1,3) +
      rec_sig(1,4) rec_sig(1,5)+rec_sig(1,6)+rec_sig
      (1,7)+rec_sig(1,8) rec_sig(1,9)+rec_sig(1,10)+
      rec_sig(1,11)+rec_sig(1,12)];//detection of
      information from received signal
62 final_sig=(1/4)*det_sig;
63 disp('transmited information is');
64 disp(final_sig)//information transmmited using
      selected valid detection code
65 //input a=4
66 /W=[0 0 0 0 ; 0 1 0 1; 0 0 1 1 ; 0 1 1 0]
67 / \text{detection code} = 2, output=-1 1-1 \text{(information of }
      user 2 spreaded with second row of Walsh Matrix)
68 //detection code > 3, results : code not available
      0 0 0
```

# PN sequence generation.

Scilab code Solution 6.1 3 bit PN sequence generation and spreading and despreading using PN sequence and shifted PN sequence

```
1 // Spreading of sequence using PN sequence and
      despreading of sequence using PN sequence and
      shifted PN sequence
2 clc;
3 clear;
4 xdel(winsid());
5 // Generation of 7 bit PN sequence
6 // Coefficient of polynomial
7 a1=1;
8 a2=1;
9 a3=1;
10 // Initial states of flip flop
11 R(1)=1;
12 R(2) = 0;
13 R(3) = 0;
14 m = 3;
15 disp('output after every clock pulse');
16 for i=1:((2^m)-1)/shift of bit in each register for
       every clock pulse
17
       r1=R(1);
```

```
18
       r2=R(2);
19
       r3=R(3);
20
       PN(i) = R(3);
       // if (a1 == 0)
21
22 R1=bitxor(r2,r3);//input of register is modulo2
      addition of R2 and R3
23 R(3) = R(2);
24 R(2) = R(1);
25 R(1) = R1;
26
27 disp(R);
28 end
29 disp('PN sequence is');
30 disp(PN); // Display 7 bit PN sequence
31 c1=[1 -1 -1]; //information of user 1
    for j=1:1:length(PN)//0 replaced with -1 in PN
32
       sequence
           if PN(j) == 0 then
33
                PN(j) = PN(j) - 1;
34
35
            else PN(j) = PN(j) + 0;
36
                 end
37
38
        end
39
        disp(PN);
40 spreaded_sig=[c1(1).*PN' c1(2).*PN' c1(3).*PN']//
      Spreading of data of user 1 using PN sequence
41 detect_code=[spreaded_sig(1:7).*PN' spreaded_sig
      (8:14).*PN' spreaded_sig(15:21).*PN'];//at
      receiver, recieved spreaded signal multiplied
      with PN sequnce
42 corr_code=[sum(detect_code(1:7)) sum(detect_code
      (8:14)) sum(detect_code(15:21))];
43 rec_sig=(1/7).*corr_code;//get information form
      received signal
44 disp('received signal with correct PN sequence is');
45 disp(rec_sig);//received data of user 1 at receiver
      :1 -1 -1
46 // Despreading with shifted PN sequence
```

```
47 shift_fact=input('enter the shifting factor');
48 \quad 1 = 1;
       k=shift_fact-1;
49
50 for i=1:1:length(PN)
                             //generation of shifted PN
      sequence as per entered shifting factor
           i<=shift_fact
51
52
             shift_seq(i)=PN(length(PN)-k);
53
             k=k-1;
             else i>shift_fact
54
             shift_seq(i)=PN(1);
55
             1=1+1;
56
57
         end
58
        end
        disp('shifted sequence is');
59
        disp(shift_seq');//display shifted sequence
60
    //despreading using shifted PN sequence
61
62 detect_shift_code=[spreaded_sig(1:7).*shift_seq'
      spreaded_sig(8:14).*shift_seq' spreaded_sig
      (15:21).*shift_seq'];
63 corr_shift_code=[sum(detect_shift_code(1:7)) sum(
      detect_shift_code(8:14)) sum(detect_shift_code
      (15:21))];
64 rec_shift_sig=(1/7).*corr_shift_code;
      disp("recieved signal with shifted PN sequence is
65
66
      disp(rec_shift_sig); // Invalid data received
         beacuse signal was despreded with shifted PN
         sequence
67
      disp('which is not valid transmitted signal');
68
      //Result:
      //output of PN sequence generator after each
69
         clock pulse
     // PN =0 0 1 0 1 1 1 replace 0 with -1,PN=-1 -1 1
70
        -1 \ 1 \ 1 \ 1
     //entered shifting factor =3, shifted PN sequence=
71
          1 \quad 1 \quad 1 \quad -1 \quad -1 \quad 1 \quad -1
72
     //Invalid signal is received when despreading is
        with shifted version of PN
```

73 // rec\_shift\_sig=- 0.1428571 0.1428571 0.1428571

# Equalization.

Scilab code Solution 7.1 Adaptive equalization using LMS filter

```
1 // Least Mean Square adaptive equalizer
2 clc;
3 clear all;
4 xdel(winsid());
5 \text{ numPoints} = 500;
6 numTaps = 1; //channel order
7 \text{ Mu} = 0.01;
                       //iteration step size
9 // input is guassian
10 x = rand(numPoints,1,'normal') + %i*rand(numPoints
     ,1, 'normal');
11 //choose channel to be random uniform
12 h = rand(numTaps,1) + %i*rand(numTaps, 1);
14 h = h/max(abs(h)); //normalize channel
15 // convolve channel with the input
16 d = filter(h, 1, x);
17
18 //initialize variables
19 w = [];
20 y = [];
```

```
21 \text{ in = []};
22 e = []; // error, final result to be computed
24 \text{ w} = \text{zeros}(\text{numTaps}+1,1) + \text{%i*zeros}(\text{numTaps}+1,1);
25 \text{ kk=1};
26 aa(kk,:)=w';
27 //LMS Adaptation
28 \text{ for } n = \text{numTaps+1} : \text{numPoints}
29
30
       // select part of training input
        in = x(n : -1 : n-numTaps);
31
       y(n) = w' * in;
32
33
34
       // compute error
       e(n) = d(n)-y(n);
35
36
       // update taps
37
38
39 w = w + Mu*(real(e(n)*conj(in)) - %i*imag(e(n)*conj(
      in)));
40
41 kk=kk+1;
42 aa(kk,:)=w';
43 end
44
45 // Plot results
46 figure;
47 iter=1:500
48 plot2d(iter,abs(e),5,logflag="nn");
49 title(['LMS Adaptation Learning Curve Using Mu =
      0.01;
50 xlabel('Iteration Number');
51 ylabel('Output Estimation Error in dB');
52 figure;
53 plot3d(abs(aa(:,1)),abs(aa(:,2)),abs(e));
54 title('LMS adaption curve with weight factors');
55 xlabel('adaptive weight factor1');
56 ylabel('adaptive weight factor2');
```

```
57 zlabel('mean square error');
58 // Output shows plot of MSE with no. of iterations
    in figure 1 and 3D plot of MSE with weight
    factors
```

## Experiment: 8

# Channel Coding using Linear Block Code

Scilab code Solution 8.1 Linear Block Coding over AWGN channel

```
1
3 //this is a linear block coding and decoding over
     awgn channel
4 // 4 bits input signal is coded with linear block
     code (4,7), 7 bit coded signal is transmitted
     over awgn channel and at receiver side signal is
     decoded. If there is error in one bit, li//near
     block code correct that error and original
     transmitter code is receved.
5 // If error is in more than one bit, code is not
     corrected so wrong code is recieved
6 clc;
7 clear all;
8 xdel(winsid());
9 global P n k;
10
11 n=7; //length of coded input
12 k=4; //length of input
```

```
13 P=[1 1 0; 0 1 1; 1 0 1;1 1 1]; //parity matrix of
      size k*(n-k) to be
14 //
                                      selected so that
     the systematic generator
                                      matrix is linearly
15 //
     independent or full rank
16 //
                                      matrix
17
18 //(n,k) linear block code where k - no. of input
      data bits and n-no. of o/p
19 //data bits. code rate=k/n
20 // x is an input vector containing k bits
21
22 //This is an linear block encoding function
23 function y1=linblkcode(x);
24 global P n k;
25 n = 7;
26 k=4;
27 P=[1 1 0; 0 1 1; 1 0 1;1 1 1]; // parity matrix
28 / x = [0 \ 1 \ 1 \ 0];
29
                    % Generator matrix k*n
30 //G=[ ];
31 G = [eye(k,k) P];
32
33 \ y1=zeros(1,n);
34 for i=1:k//linear block coding
35
       var(i,:)=x(1,i) & G(i,:);
       var(i,:)=bool2s(var(i,:));
36
       y1(1,:)=bitxor(var(i,:),y1(1,:));//coded signal
37
38 end
39
40 endfunction
41
42
43 //%This is a linear block syndrome decoding function
       file%
44
45 function x1=linblkdecoder(y)
```

```
46 //% here y is recieved vector 7 bits long
47
48 //\% (7,4) linear block code
49 global P n k;
50
51
52 //H=[ ]; //% PARITY CHECK MATRIX
53
54 H = [P' eye((n-k), (n-k))];
55 Ht=H'; //%transpose of H
56
57 S=zeros(1,n-k); //%syndrome of recieved vector x
58 for i=1:n-k// decoding of linear block code
       S(i)=y(1) & Ht(1,i);
59
       S(i) = bool2s(S(i));
60
61
       for j=2:n
62
            S(i)=bitxor(S(i), bool2s((y(j) & Ht(j,i))));
63
               //decoded signal
64
        end
65 end
66
67
68
69
   //%%****SYNDROME LOOK UP TABLE********
70
71
   //%%*************
72 / \%
73 \text{ if } S == [0 \ 0 \ 0]
      e = [0 \ 0 \ 0 \ 0 \ 0 \ 0];
74
       z=bitxor(y,e);
75
76 end
77
78 \text{ if } S == [0 \ 0 \ 1]
      e = [0 \ 0 \ 0 \ 0 \ 0 \ 1];
79
       z=bitxor(y,e);
80
81 end
82 \text{ if } S == [0 \ 1 \ 0]
```

```
e = [0 \ 0 \ 0 \ 0 \ 1 \ 0];
 83
 84
         z=bitxor(y,e);
 85 end
 86 \text{ if } S == [1 \ 0 \ 0]
        e = [0 \ 0 \ 0 \ 0 \ 1 \ 0 \ 0];
 87
 88
         z=bitxor(y,e);
 89 end
90 \text{ if } S == [1 \ 1 \ 1]
        e=[0 0 0 1 0 0 0];
 91
 92
         z=bitxor(y,e);
93 end
94 \text{ if } S == [1 \ 0 \ 1]
95
        e = [0 \ 0 \ 1 \ 0 \ 0 \ 0];
         z=bitxor(y,e);
96
97 end
98 \text{ if } S == [0 \ 1 \ 1]
        e=[0 1 0 0 0 0];
99
         z=bitxor(y,e);
100
101 end
102 \quad if \quad S == [1 \quad 1 \quad 0]
103
        e = [1 \ 0 \ 0 \ 0 \ 0 \ 0];
104
         z=bitxor(y,e);
105 end
106 // disp ('error');
107 // disp (e);
108 x1=z(1,1:k);
109 endfunction
110 snr_dB=2;
111
112 x = [1 \ 0 \ 0 \ 1]; //
                                              input bits to the
        encoder of size 1* k
113 y1=linblkcode(x);//
                                          //
                                               y1 is the output
        of linear block encoder
114 n1 = 1/sqrt(2)*[rand(1,length(y1),'normal') + %i*
       rand(1,length(y1), 'normal')];//white gaussian
        noise generation
115 r=y1+ 10^{-(-snr_dB/20)*n1}; //received signal over awgn
         channel
```

```
116 / r1 = real(r)
117 rec=real(r) \ge 0.5; // detection of bit 1 and 0 in
      received signal
118 rec_fin=bool2s(rec);//convert boolean matrix to zero
       one matrix
119 / rec_err = rec_fin = y1;
120 //no_err=bool2s(rec_err);
121 disp('The information signal=')//display input
122 \quad disp(x)
123 disp('The transmitted encoded signal=')//display
      coded signal
124 disp(y1)
125 disp('The recieved signal=')//display received
      signal
126 disp(rec_fin);
127 x1=linblkdecoder(rec_fin); //
                                             \% x1 is the
      output of the linear block decoder
128 disp('The decoded signal=')//display decoded signal
129 disp(x1);
130 if x1==x then disp('one or less than one error so
      correct code is received');
131 else
132
        disp ('more than one error so wrong code detected
           ');
133 end
134 //Output: The information signal is : 1001
135 //transmitted code is : 1001001
136 //1. received signal is :1011001(e.g)(error in only
      one bit)
137 //decoded signal: 1001
138 //one or less than one error so correct code is
      received
139 //2. received signal is:1011011(e.g)(error in more
      than one bits)
140 //decoded signal:1010
141 //more than one error so wrong code is received
```

# Experiment: 9

#### Transmit and receive diversity

Scilab code Solution 9.1 Selection Diversity over AWGN channel

```
1 //ber performance with 1, 2 and 3 receiver antennas
      over awgn channel using selection diversity
2 clc;
3 clear;
4 xdel(winsid());
5 sym=10000; //no. of symbols
6 data1=grand(1,sym,"uin",0,1);//randomly generated
7 s = 2*data1-1; // BPSK modulation 0 -> -1; 1 -> 1
8 nRx = [1 \ 2 \ 3]; //no .of receiving antennas
9 \text{ snr_dB} = [1:10]; // \text{ signal to noise ratio}
10 for j = 1:length(nRx)
       for i = 1:length(snr_dB)
11
12
           n = 1/sqrt(2)*[rand(nRx(j), sym, 'normal') +
              %i*rand(nRx(j),sym,'normal')]; //white
              gaussian noise
13
14
           y = ones(nRx(j),1)*s + 10^{-snr_dB(i)/20}*n;
              //received signal over awgn channel
            [yHat1 ind] = mtlb_max(y,[],1); //find
15
               strongest received signal from all
```

```
antennas
16
17
           ipHat1 = real(yHat1)>0;
           ipHat = bool2s(ipHat1); //boolean to zero one
18
               matrix conversion
19
           // effective SNR
20
            nErr(j,i) = size(find([data1 - ipHat]),2);//
               no. of error calculation
21
               end
22 end
23 simBer = nErr/sym; //BER calculation
24 // plot of ber comparison plot for 1,2 and 3
      receiving antennas
25 snr_dB=1:10
26 plot2d(snr_dB, simBer(1,:),5,logflag="nl");
27 plot2d(snr_dB, simBer(2,:),2,logflag="nl");
28 plot2d(snr_dB, simBer(3,:),12,logflag="nl");
29 xgrid
30 legend(['1X1';'1X2';'1x3']);
31 xlabel('Number of receive antenna');
32 ylabel('effective SNR, dB');
33 title('SNR improvement with Selection Combining');
34 //output presents BER performance comparison plots
      with 1,2 and 3 receiving antennas over awgn
      channels
```

Scilab code Solution 9.2 Maximal Ratio Combining over AWGN and Rayleigh fading Channel

```
1 // BER Performance coamparison with one receiviving
     atenna and two receiving antennas with Maximal
     ratio Combining diversity technique over awgn
     channe and rayleigh fading channel
2 clc;
3 clear;
```

```
4 xdel(winsid());
5 \text{ sym} = 100000; // \text{ no. of symbols}
7 data1=grand(1,sym,"uin",0,1);// input signal is
      randomly generated
8 / N = 10; % number of bits or symbols
9 //ip = rand(1,N) > 0.5; % generating 0,1 with equal
      probability
10 s = 2*data1-1; // BPSK modulation 0 \rightarrow -1; 1 \rightarrow 1
11 nRx = [1 \ 2]; //no \ of receivers
            [1:20]; // signal to noise ration in dB
12 \text{ snr_dB} =
13 for jj = 1:length(nRx)
14
       for ii = 1:length(snr_dB)
15
           n = 1/sqrt(2)*[rand(nRx(jj),sym,'normal') +
              %i*rand(nRx(jj),sym,'normal')]; //white
              gaussian noise,
           h = 1/sqrt(2)*[rand(nRx(jj),sym,'normal') +
16
              %i*rand(nRx(jj),sym,'normal')]; //
               Rayleigh fading channel
17
           // Channel and noise Noise addition
           sD = kron(ones(nRx(jj),1),s);
18
           y = h.*sD + 10^(-snr_dB(ii)/20)*n;//
19
              received signal over awgn channel and
               ayleigh fading channel
20
           // finding the power of the channel on all
              rx chain
21
                     sum(conj(h).*y,1)./sum(h.*conj(h)
                ,1); // maximal ratio combining
          // hPower1 = h.*conj(h);
22
23
24
           ipHat = real(yHat)>0;
25
           // effective SNR
                    nErr(jj,ii) = size(find([data1-
26
                       ipHat]),2);//calculate error
27
                end
28 end
29 simBer = nErr/sym; //bit error rate calculation
30 // plot
```

### Experiment: 10

# Speech coding

Scilab code Solution 10.1 speech coding and Decoding using LPC

```
1 //Exp-10 Speech coding using Long Term Predictive
     coder
2 //This code read way file and play original signal
     and compressed signal
3 // It also plots original signal as well as
     compressed signal
5 function [aCoeff, tcount_of_aCoeff, e] =
     func_lev_durb(y, M);
6 //M=order and y is array of the data point of the
     current frame
               //initializing summartion term "sk"
7 \text{ sk=0};
8 a=[zeros(M+1);zeros(M+1)]; //defining a matrix of
      zeros for "a" for init.
  //MAIN BODY OF THIS PROGRAM STARTS FROM HERE
     >>>>>>>>
10 z = x corr(y);
11
12 //finding array of R[l]
13 R=z( ((length(z)+1) ./2) : length(z)); //R=array
     of "R[1]", where l = 0, 1, ...(b+N)-1 %R(1)=R[lag
```

```
=0], R(2)=R[lag=1], \Re R(3)=R[lag=2]... etc
14
  //GETTING OTHER PARAMETERS OF PREDICTOR OF ORDER
15
      " 0":
16 s = 1;
               //s = step no.
                        //J=array of "Jl", where l
17 J(1) = R(1);
      =0,1,2...(b+N)-1, J(1)=J0, J(2)=J1, J(3)=J2 etc
18
  //GETTING OTHER PARAMETERS OF PREDICTOR OF ORDER "(s
      -1)"
20 \text{ for } s=2:M+1,
                            //clearing "sk" for each
21
       sk=0;
          iteration
22
       for i=2:(s-1),
23
           sk = sk + a(i,(s-1)).*R(s-i+1);
                            //now we know value of "sk",
24
       end
           the summation term
                             //of formula of calculating
25
                               "k(l)"
26
       k(s) = (R(s) + sk)./J(s-1);
27
       J(s)=J(s-1).*(1-(k(s)).^2);
28
29
       a(s,s) = -k(s);
       a(1,s)=1;
30
31
       for i=2:(s-1),
32
           a(i,s)=a(i,(s-1)) - k(s).*a((s-i+1),(s-1));
33
       end
34
       end
  //increment "b" and do same for next frame until end
       of frame when
36 //combining this code with other parts of LPC algo
37
38 //PREDICTION ERROR; FOR TESTING THE ABOVE PREDICTOR
                              //array of "a(i,s)", where
39 aCoeff=a((1:s),s)';
      , s=M+1
40 tcount_of_aCoeff = length(aCoeff);
41
    y_padded_for_delay_r = [y'; zeros(1,1)]; //it is
42
```

```
padded with zeros to remove the effect of delay
       in filter
    est_y_with_dummy_pad = filter([0 -aCoeff(2:9)],1,
43
       y_padded_for_delay_r); // = s^n(n) with a cap
       on page 92 of the book
44 \text{ est_y} = \text{est_y_with_dummy_pad}(2:321);
45 e = y' - est_y; //supposed to be a white noise
46 endfunction
47
48 function [aCoeff, b_LTopt, Topt, e_prime] =
      f_ENCODER_relp(x, fs)
49 M = 8; //prediction order for LP analysis
50 //INITIALIZATION;
51 b=1:
               //index no. of starting data point of
      current frame
52 \text{ fsize} = 20e-3;
                     //frame size (in milisec)
53 frame_length = round(fs .* fsize);
                                        //=number data
      points in each framesize of "x"
54 N= frame_length - 1; //N+1 = frame length = number
      of data points in each framesize
55
56 \text{ y_proc} = \text{filter}([1 -1], [1 -0.999], x); //pre-
      processing
57 //FRAME SEGMENTATION
58 for b=1 : frame_length : (length(x) - N)
59
                             //"b+N" denotes the end
60 \text{ y_f} = \text{y_proc(b:b+N)};
      point of current frame. "y" denotes an array of
      the data points of the current frame
    //LP ANALYSIS [lev-durb] & PREDICTION ERROR (short-
61
       term) FILTER;
62
       [a, tcount_of_aCoeff, e_s] = func_lev_durb (y_f,
           M); //e=error signal from lev-durb proc
       aCoeff(b: (b + tcount_of_aCoeff - 1)) = a;
63
          aCoeff is array of "a" for whole "x"
       //LONG-TERM LP ANALYSIS, FILTERING, AND CODING
64
          analysis:
       T_{min} = round (fs .* 5e-3); //=total data
65
```

```
points in 5ms of "x"
       T_{max} = round (fs .* 15e-3);
66
67
       for bs = b : 40 : b+length(y_f)-40 //subframing
68
           bs = 1281;
69
            if bs < T_max</pre>
70
                break;
71
            end
72
73
            Jmin(bs) = 10^9;
74
75
            for T = T_min : T_max
                                         // within 1 (
               current) frame T = 40;
                    for c = 1:40
                                     //data points of
76
                       current subframe c=1; temporary
                         sm1(c) = (y_proc(bs+(c-1)) .*
77
                            y_{proc}(bs-T+(c-1)); //es(n)
                         sm2(c) = y_proc(bs-T+(c-1)); //=
78
                            es (n-T)
                         sm22(c) = sm2(c).^2;
79
80
                    end
81
                    q1 = sum(sm1);
                    q2 = sum(sm22);
82
                    b_{LT}(T) = -(q1./q2);
83
                //J loop:
84
                    for c = 1:40 //data points of
85
                       current subframe c=1; temporary
                         smJ1(c) = y_proc(bs+(c-1));
86
                         smJ2(c) = b_LT(T) .* y_proc(bs-T)
87
                            +(c-1));
88
                    end
89
                    smJ = smJ1 + smJ2;
90
                    qJ = smJ.^2;
                    J(T) = sum(qJ);
91
92
                    if J(T) < Jmin(bs),</pre>
93
94
                         Jmin(bs) = J(T);
                         Topt(bs) = T;
95
```

```
96
                         if b_LT(T) >= 1,
97
                             b_LTopt(bs) = 0.9999; //
                                 trancation
98
                         else
99
                             b_LTopt(bs) = b_LT(T);
100
                         end
101
                     else
102
                    end
                    //T loop ends
103
            end
            //predictor:
104
            LT_gain = [zeros(1, Topt(bs)-1), b_LTopt(bs)
105
                      //as it says z^-T in page 121
106
             e_s_padded_for_delay_r = [e_s(c1:c1+39);
                zeros(Topt(bs), 1)]; //it is padded with
                 zeros to remove the effect of delay in
                filter. %Topt(bs) no. of 'z's and one
                '1' results in total 'Topt(bs)' amount
                of delay
            e_with_dummy_pad = filter([1 LT_gain], 1,
107
               e_s_padded_for_delay_r); // = 1 + 0*z
               ^{-1} + 0*z^{-2} + \dots + b*z^{-T}
             e_LT(bs:bs+39,1) = e_with_dummy_pad(Topt(bs
108
                )+1 : Topt(bs)+1+39); //LT predicted "
                e"
            e(bs:bs+39, 1) = e_s(c1 : c1+39) - e_LT(bs : c1+39)
109
                bs + 39);
110
            //WEIGHTING FILTER:
111
112
            w = [-0.0004;
               -0.0156; -0.0677; 0.0545; 0.6069; 1.0000; 0.6069; 0.0545; -0.067
                 //11 point flattop window is
               temporarily chosen
            wndd = conv(w, e(bs:bs+39));
                                                //outputs
113
               total 50 samples
            x_n(bs:bs+39) = wndd(6:45); //middle 40
114
               samples are taken
115
            //POSITION SELECTION & EXCITATION GENERATOR:
116
```

```
117
            for i1 = 0:3
118
                for i = i1+bs : 3 : bs+i1+38;
                    x_m(i1+1,i) = x_n(i);
119
120
                end
121
122
                E_m(i1+1,1) = sum((x_m(i1+1, bs:3)).^2);
123
            end
            [E_m_max, index_max] = gsort(E_m);
124
            e_{prime}(bs : bs+39) = x_m(index_max(4), bs:
125
               bs + 39);
            c1 = c1 + 40;
126
127
        end
128 end
129 endfunction
130
131 //RELP DECODER portion:
132 function [synth_speech, synth_speech1, LT_gain,
      e_prime_pad_for_d_r, e_prime_op_dummy_pad,
      e_prime_op, e_prime_op_pad_delay_r,
      synth_speech_dummy_pad] = f_DECODER_relp(aCoeff,
      b_LTopt, Topt, e_prime)
133 //re-calculating frame_length for this decoder
134 frame_length=9; //initial value for calculation
135 for i=10:length(aCoeff)
        if aCoeff(i) == 0
136
137
            frame_length = frame_length + 1;
138
        else break;
139
        end
140 end
                         //making it a column matrix
141 e_prime = e_prime';
      for convenience
142
143 for b=1 : frame_length : length(aCoeff)
                                                //length(
      aCoeff) should be very close (i.e less than a
      frame_length error) to length(x)
        for bs = b : 40 : b+frame_length-40
144
           subframing
145
```

```
//EXCITATION GENERATOR: not done yet.
146
               because e_prime has been sent to this
               decoder directly, without quantization.
            //PITCH SYNTHESIS FILTER:
147
                                          %has to be done
                per subframe
            LT_gain = [zeros(1, Topt(bs)-1), b_LTopt(bs)
148
                      //as it says z^-T
             e_prime_pad_for_d_r = [e_prime(bs:bs+39);
149
                zeros(Topt(bs), 1)]; //it is padded with
                 zeros to remove the effect of delay in
                filter. %Topt(bs) no. of 'z's and one
                '1' results in total 'Topt(bs)' amount
                of delay
150
            e_prime_op_dummy_pad = filter(1, [1 LT_gain
               ], e_prime_pad_for_d_r); //= 1 / (1 +
               0*z^{-1} + 0*z^{-2} + \dots + b*z^{-T}
            e_prime_op(bs:bs+39,1) =
151
               e_prime_op_dummy_pad(Topt(bs)+1 : Topt(bs
               )+1+39);
                          //pitch-synthesis filter
               output
                   //FORMANT SYNTHESIS FILTER:
152
        end
           e_prime_op_pad_delay_r= [e_prime_op(b : b
153
              +159); zeros(1,1)]; //it is padded with
              zeros to remove the effect of delay in
154
            synth_speech_dummy_pad = filter(1, [1 aCoeff
               (b+1 : b+8)], e_prime_op_pad_delay_r);
            synth_speech1(b : b+159) =
155
               synth_speech_dummy_pad (2:161);
               EMPHASIS (de-proprocessing):
            synth_speech(b : b+159) = filter([1 -0.999],
156
                [1 -1], synth_speech1(b : b+159)); //De
               -processing
157 end
158 endfunction
159
160 clc;
161 clear all;
```

```
162 xdel(winsid());
163 inpfilenm = "SCI/modules/sound/demos/slofwb.wav";
164 [x,fs,bits] = wavread(inpfilenm);
165
166 t=length(x)./fs;// total time t seconds
167 //COMPRESSION STARTS HERE,
168 disp('original signal');
169 sound(x, fs);
170 [aCoeff, b_LTopt, Topt, e_prime] = f_ENCODER_relp(x,
171
172
            // e_prime is instead of position,
               peak_magitude_index and
               sample_amplitude_index. (temporarily)
173 //halt()
174 //halt ('Press a key to play the original sound!')
175
    [synth_speech] = f_DECODER_relp(aCoeff, b_LTopt,
176
      Topt, e_prime);
177
178 //RESULTS,
179
180
181 disp('compressed signal');
182 sound(synth_speech, fs);
183
184 figure;
185 subplot (211),
186 plot(x); title(['Original signal = "', inpfilenm, '"
       <sup>'</sup>]);
187 subplot(212), plot(synth_speech); title('RELP
       compressed output');
188 //Output plays original signal and after
      approximately 5 minutes it plays compressed sound
       and plot the original signal and compressed
      signal.
```