B.M.S COLLEGE OF ENGINEERING

(Autonomous College under VTU)

Department of Telecommunication Engineering

(Accredited by NBA for SIX years under Tier - I format)



DIGITAL COMMUNICATION 16TE6DCDCM

REPORT ON

ADAPTIVE DELTA MODULATION WITH DUO BINARY ENCODING

(AN END TO END DIGITAL COMMUNICATION PROJECT)

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Certificate

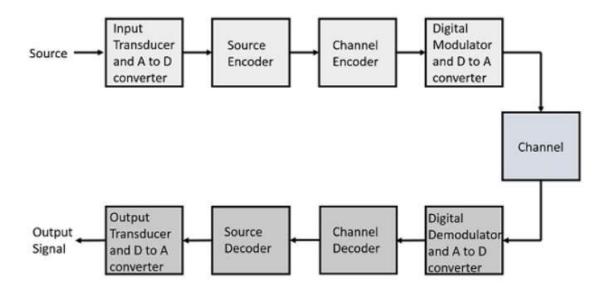
This is to certify that the self-study entitled "ADAPTIVE DELTA MODULATION WITH DUO BINARY ENCODING" is submitted by ABHASH KUMAR JHA (USN:1BM17TE-002), ADVAITH SHANKAR (USN:1BM17TE-004), ANUKRITI SHARMA (USN: 1BM17TE-009), SHUBHA PRASAD (USN: 1BM14TE0-052) in partial fulfilment for the subject of DIGITAL COMMUNICATION during VI semester of Telecommunication Engineering of academic year 2020-2021

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1. Introduction to Communication Model

The communication that occurs in our day-to-day life is in the form of signals. These signals, such as sound signals, generally, are analog in nature. When the communication needs to be established over a distance, then the analog signals are sent through wire, using different techniques for effective transmission.

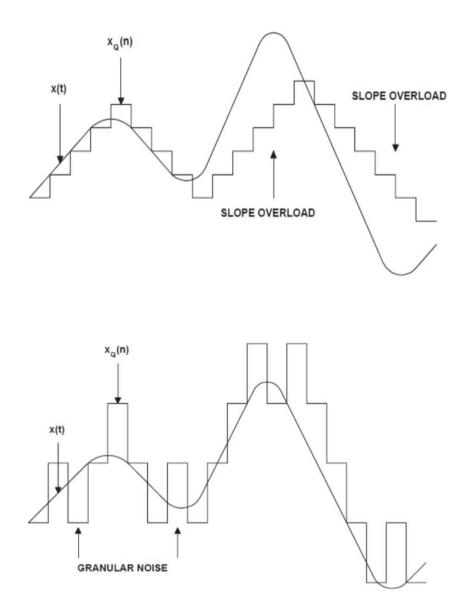


In this project, we are going to implement a Transmitter-Channel-Receiver Model in MATLAB and try to reciprocate the noise into the system for practical view of the system.

We are going to implement Adaptive Delta Modulation Technique and a correlative coding i.e. Duobinary Signaling.

2. Adaptive Delta Modulation

This Modulation is the refined form of delta modulation. This method was introduced to solve the granular noise and slope overload error caused during Delta modulation.



This Modulation method is similar to Delta modulation except that the step size is variable according to the input signal in Adaptive Delta Modulation whereas it is a fixed value in delta modulation

2.1 Working

There are several types of ADM, depending on the type of scheme used for adjusting the step size. In this project, the implementation for step-size is based upon the SONG Algorithm. In practical implementations of the system, the step size $\Delta(nTs)$ or $2\partial(nTs)$ is constrained to lie between minimum and maximum values.

The processor detects the pattern to see if the delta modulator is operating in the granular noise region, in which case it produces an alternating1010.....

Pattern, or in the slope over load region in which case it produces an all-1 or all-0 pattern.

- If the ADM senses a ...1010.... pattern, it decreases the step-size, and
- If it senses1111.... or0000.... pattern, it increases the step-size.

Let m(t) be the input signal and be its staircase approximation. Let error, at the kth sampling instant (k = 0, 1, 2, 3...) be e(k) which can be either positive, negative or zero. Then,

$$k^{th} \ bit = \begin{cases} 1 & e(k) > 0 \\ 0 & e(k) < 0 \\ 0 \ or \ 1 & e(k) = 0 \end{cases}$$

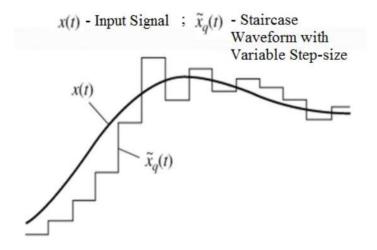
The SONG algorithm used by NASA produces the step-size $\partial(k+1)$ which minimizes the mean-square error between m(t) and m^(t). In the implementation of the SONG system ± 5 V was the maximum signal level and the minimum step-size was So = 10 mV.

Here we see that as long as e(k) is of the same sign as e(k-1) the magnitude of the new step-size $\partial(k+1)$ will exceed the magnitude of the old step-size $\partial(k)$ by So, the minimum step-size. However, if e(k) and e(k-1) differ in sign, the magnitude of $\partial(k+1)$ will be less than the magnitude of $\partial(k)$ by the amount So. The algorithm can also be written in terms of the following equation:

$$|\partial(k+1)| = \begin{cases} |\partial(k)| + \partial(0) & \text{if } e(k) = e(k-1) \\ |\partial(k)| - \partial(0) & \text{if } e(k) \neq e(k-1) \end{cases}$$

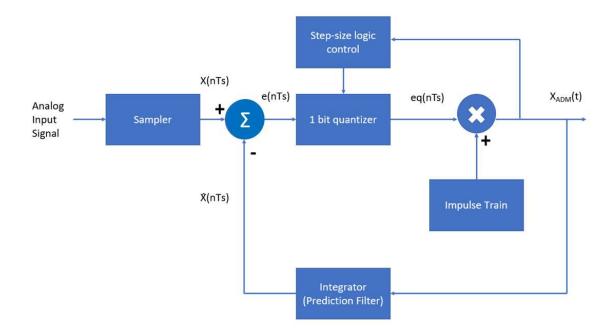
A more general form for the same equation will be,

$$\partial(k) = |\partial(k)|.sgn(e(k)) + \partial(0).sgn(e(k-1))$$

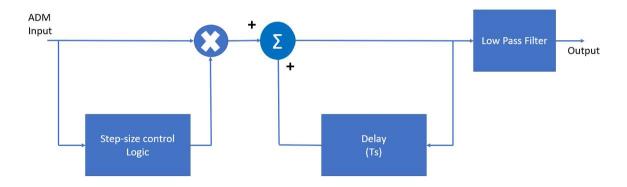


2.2 Block Diagram

TRANSMITTER:



RECIEVER:



2.3 Applications

- This modulation is used for a system which requires improved wireless voice quality as well as speed transfer of bits.
- In television signal transmission this modulation process is used.
- This modulation method is used in voice coding.
- This modulation is also used as a standard by NASA for all communications between mission control and spacecraft.
- Motorola's SECURENET line of digital radio products uses 12kbits/sec Adaptive Delta Modulation.
- To provide voice detection quality audio at deployed areas, military uses 16 to 32 kbit/sec modulation system in TRI-TAC digital telephones.
- US army forces use 16kbit/sec rates to conserve bandwidth over tactical links.
- For improved voice quality US Air Forces uses 32kbits/sec rates.
- In Bluetooth-services to encode voice signals, this modulation is used with 32bits/sec rates.

3. Duobinary Encoder/Decoder

The ISI is treated as an undesirable phenomenon that produces a degradation in system performance, but by adding ISI to the transmitted signal in a controlled manner, it is possible to achieve a bit rate of 2Bo bits per second in a channel of bandwidth Bo Hz. Such a scheme is **correlative coding** or **partial-response signalling** scheme.

One such example is **Duo binary signalling**. Duo means transmission capacity of system is doubled.

3.1 Working

Precoding:

In case of duo binary coding if error occurs in a single bit it reflects as multiple errors because the present decision depends on previous decision also. To make each decision independent we use a precoder at the receiver before performing duo binary operation.

The precoding operation performed on the input binary sequence $\{b_k\}$ converts it into another binary sequence $\{a_k\}$ given by:

$$a_k = b_k \oplus a_{k-1}$$

This $\{a_k\}$ stream is then converted into appropriate NRZ scheme. For simplicity, this project is using binary scheme i.e.

$$a_k = \begin{cases} -1, & \{a_k\} = 0 \\ 1, & \{a_k\} = 1 \end{cases}$$

Encoding:

The Encoding is done by adding the current bit with the previous bit i.e. the delayed element of the same bit.

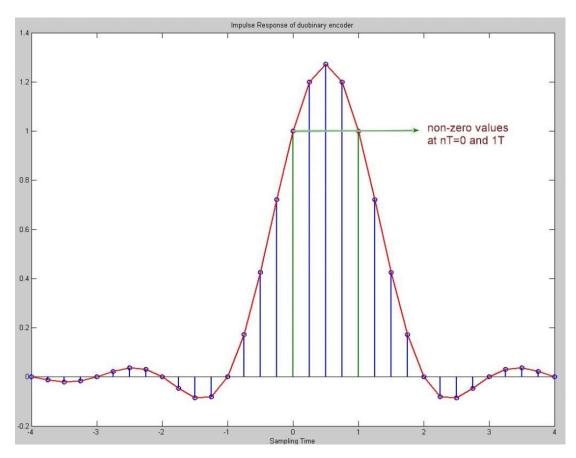
$$c_k = a_k + a_{k-1}$$

If that symbol at precoder is in polar format c_k takes three levels,

$$\mathbf{C_k} = \begin{cases} \pm 2\mathbf{v} & \text{if} & \mathbf{b_k} = \text{ symbol } 0 \\ 0\mathbf{v} & \text{if} & \mathbf{b_k} = \text{ symbol } 1 \end{cases}$$

From the diagram, impulse response of the duobinary encoder is computed as:

$$h(t) = sinc\left(rac{t}{T}
ight) + sinc\left(rac{t-T}{T}
ight)$$

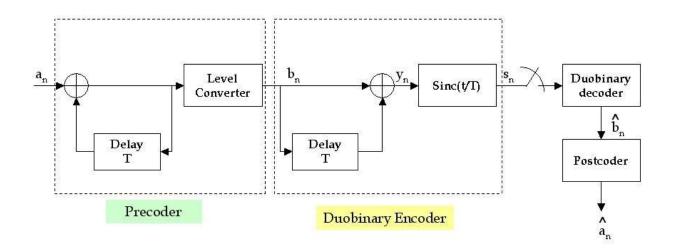


Decoding:

The decision rule for detecting the original input binary sequence $\{b_k\}$ from $\{c_k\}$ is:

$$\widehat{b_k} = \begin{cases} 1, & |c_k| < 1V \\ 0, & |c_k| \ge 1V \end{cases}$$

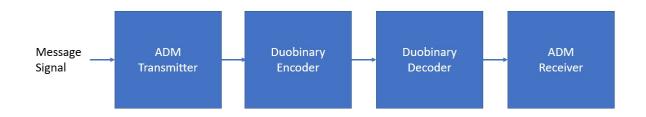
3.2 Block diagram



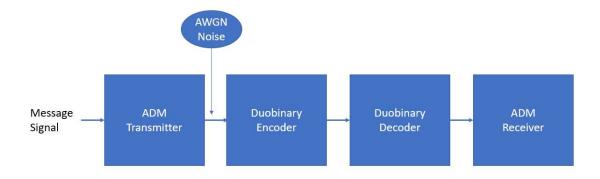
Today the duo-binary techniques are widely applied throughout the world. While all current applications in digital communications such as data transmission, digital radio, and PCM cable transmission, and other new possibilities are being explored. This technique has been applied to fiber optics and to high density disk recording which have given excellent results

4. Meta Block Diagram

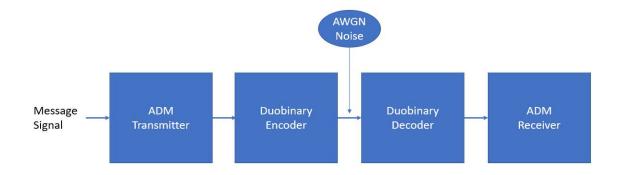
The block diagram for the system without noise will be:



The block diagram for the system with noise at the transmitter's staircase output will be:



The block diagram for the system with noise in between Duobinary Encoder and Decoder is:



<u>5. Code</u>

Structure of the code:

Functions:

ad_deltamod.m	ADM Transmitter
ad_deltademod.m	ADM Receiver
Duobinary_Encoder.m	Duobinary Encoder
Duobinary_Decoder.m	Duobinary Decoder

Runner files:

- Communication_system.m
- Communication_system_duobinary_error.m

ad deltamod.m

```
function [c,xq,zero flag array] = ad deltamod(x,delta min)
%This is a Adaptive delta Modulator
응 {
x -> The signal input, a vector
delta min -> min step size
zero flag array -> zero indicator to avoid zero error
c -> encoded digital output
응 }
   N = length(x);
   delta_array = zeros(1,N); %variable delta
   e = zeros(1,N); %error
eq = zeros(1,N); %quantized error
   xq = zeros(1,N);
   %zero flag
    zero flag array = zeros(1,N);
    for i = 1:N
        if x(i) == 0
            zero_flag_array(i) = 1;
        end
    end
    for i = 1: length(x)
            if (i==1)
                e(i) = x(i);
                if x(i) == 0
                    delta array(i) = delta min;
                    delta array(i) = delta min*sign(e(i));
                end
                eq(i) = delta array(i);
                xq(i) = eq(i);
            else
                e(i) = x(i) - xq(i-1);
                delta array(i) = (abs(delta array(i-1))*sign(e(i))) +
(delta array(1)*sign(e(i-1)));
                %delta array(i) = delta min*sign(e(i));
                eq(i) = delta array(i);
                xq(i) = eq(i) + xq(i-1);
            end
    end
    c = ones(1,N);
    for i = 1:N
        if e(i) > 0
            %only positive vals
            c(i) = 1;
            %-ve + 0
            c(i) = 0;
```

```
end
end
```

end

ad_deltademod.m

```
function [rec xq, signal] = ad deltademod(c, zero flag array, delta min)
%This is a adaptive delta demodulator
L = length(c);
signal = zeros(1,L);
rec_delta_array = zeros(1,L);
sign_array = zeros(1,L);
rec xq = zeros(1,L);
%first we need to generate our error/sign array
for i = 1:L
    if c(i) == 0
       sign_array(i) = -1;
        sign_array(i) = 1;
    end
    if zero_flag_array(i) == 1
        sign_array(i) = 0;
end
for i = 1:L
        if (i==1)
            if(sign array(i) == 0)
                rec_delta_array(i) = delta_min;
                rec_delta_array(i) = delta_min*sign_array(i);
            end
            rec_xq(i) = rec_delta_array(i);
            signal(i) = rec xq(i);
        else
            rec_delta_array(i) = (abs(rec_delta_array(i-1))*sign_array(i)) +
(rec_delta_array(1)*sign_array(i-1));
            rec_xq(i) = rec_delta_array(i) + rec_xq(i-1);
            signal(i) = signal(i-1) + rec xq(i);
        end
end
end
```

Duobinary_Encoder.m

```
function [c] = Duobinary_Encoder(b)
%This is a precoded duobinary encoder
%b is the input binary sequence : precoded input
% a_volts is basically the NRZ encoder output
%c is the duobinary coded output
%variables
a = zeros(1,length(b));
```

```
a volts = zeros(1,length(b));
a(1) = xor(1,b(1));
%NRZ Encoder for converting bits to voltage levels
if(a(1) == 1)
   a_{volts}(1) = 1;
else
   a_{volts}(1) = -1;
end
for k = 2:length(b)
   a(k) = xor(a(k-1),b(k));
   if(a(k) == 1)
     a_{volts(k)} = 1;
   else
     a_{volts(k)} = -1;
   end
end
disp('Precoded output in binary:');
disp(a);
disp('....');
disp('Precoded output in voltage:');
disp(a volts);
disp('....');
%Encoder
c = zeros(1, length(b));
c(1) = 1 + a volts(1);
for k = 2:length(a)
  c(k) = a_volts(k-1) + a_volts(k);
```

Duobinary_Decoder.m

```
function [b_out] = Duobinary_Decoder(c)
%This is a precoded binary decoder
%c is the duobinary coded output
%b_out is the output of suobinary decoder

% decoder decision
b_out = zeros(1,length(c));
for k =1:length(c)
   if(abs(c(k)) > 1)
       b_out(k) = 0;
   else
       b_out(k) = 1;
   end
end
```

Communication_system.m

```
clear all;
close all;
clc;
%% inputs
Am = 5;
            %amplitude
                                   %input('Enter the value for amplitude');
         %frequency
fm = 1;
                                    %input('Enter the value for frequency');
fs = 30*fm; %sampling freq
                                   %input('What is the sampling frequency?');
%t = 0:1/fs:10;
dt = 1/fs;
                               % seconds per sample
StopTime = 1;
                               % seconds
t = (0:dt:StopTime-dt)';
%input signal
x = Am*sin(2*pi*fm*t);
응 {
N = size(t, 1);
y = fftshift(fft(x));
                               % hertz
dF = fs/N;
                               % hertz
f = -fs/2:dF:fs/2-dF;
\$\$Plot the spectrum:
figure;
plot(f,abs(y)/N);
xlabel('Frequency (in hertz)');
title('Magnitude Response');
delta = (2*pi*fm*Am)/fs;
%% Without Error
disp('Adaptive Delta Modulation without noise');
disp(' ');
[digital code, xq, zero track] = ad deltamod(x, delta);
%Transmittor
disp('The digital code is:');
disp(digital_code);
disp(' ');
coded = Duobinary Encoder(digital code);
%duobinary encoder
disp(' ');
disp('Encoded digital code is:');
disp(coded);
disp(' ');
decoded digital code = Duobinary Decoder(coded);
%duobinary decoder
disp('The output of duobinary decoder is');
disp(decoded_digital_code);
disp(' ');
[rec staircase,my signal] = ad deltademod(decoded digital code,zero track,delta);
%Reciever
%y = lowpass(my_signal,2*fm,fs);
%lowpass filter output
b = fir1(100, 10*fm/fs);
y = conv2(my_signal,b,'same');
%% With Error
disp('Adaptive Delta Modulation with noise');
```

```
disp(' ');
%The Transmittor part is same so, we will take same variables as above
%Adding noise to the staircase signal -> Channel
x = awgn(xq, -10);
digital code error = ones(1,length(xq));
digital code error(1) = (x error(1)>0);
for i = 2:length(xq)
   if x(i)-x error(i-1)>0
       %only positive vals
       digital code error(i) = 1;
   else
       %-ve + 0
       digital code error(i) = 0;
   end
end
disp('The noisy digital code is:');
disp(digital_code_error);
disp('....');
coded error = Duobinary Encoder(digital code error);
%duobinary encoder
disp('Encoded digital data is:') ;
disp(coded error);
disp('....');
decoded_digital_code_error = Duobinary_Decoder(coded_error);
%duobinary_decoder
disp('The output of duobinary decoder is');
disp(decoded_digital_code_error);
disp('....');
[rec staircase error, my signal error] =
ad deltademod(decoded digital code error, zero track, delta); %Reciever
%y_error = lowpass(my_signal_error,2*fm,fs);
%low pass
y error = conv2(my signal error, b, 'same');
%% Without Error Plots
%orignal signal, staircase signal, recieved staircase signal, rec low pass
figure('Name','Adaptive Delta Modulation without noise','NumberTitle','off');
plot(t,x,'DisplayName','Message signal');
title('Adaptive Delta Modulation');
xlabel('Time (in sec)');
ylabel('Amplitude (in volts)');
hold 'on';
stairs(t,xq,'DisplayName','Staircase signal');
stairs(t,rec staircase,'DisplayName','Recieved Staircase signal (Overlapping with
staircase signal of reciever)');
plot(t,y,'DisplayName','Recieved signal');
hold 'off';
legend
%Binary signals
bp = .00001; %bit period
br = 1/bp; %bit rate
digital_code_bit = [];
for n=1:1:length(digital code)
```

```
if digital code(n) == 1
       se=ones(1,100);
    else
        se=-1*ones(1,100);
    end
     digital code bit=[digital code bit, se];
end
coded bit = [];
for n=1:1:length(coded)
    if coded(n) == 2
       c se= 2*ones(1,100);
    elseif coded(n) == -2
        c se= -2*ones(1,100);
    else
        c_se = zeros(1,100);
    end
     coded bit=[coded bit c se];
end
decoded digital code bit = [];
for n=1:1:length(decoded digital code)
    if decoded_digital_code(n) == 1
       se=ones(1,100);
    else
        se=-1*ones(1,100);
    end
     decoded digital code bit=[decoded digital code bit se];
end
t1= bp/100:bp/100:100*length(x)*(bp/100);
figure('Name','Bit Transmission without error','NumberTitle','off')
subplot(3,1,1);
plot(t1,digital code bit,'lineWidth',2.5);
grid on;
axis([ 0 bp*length(coded) -1.5 1.5]);
ylabel('Amplitude(volt)');
xlabel('time(sec)');
title('Digital Signal at Transmitter');
subplot(3,1,2);
plot(t1,coded bit,'lineWidth',2.5);
grid on;
axis([ 0 bp*length(coded) -2.5 2.5]);
ylabel('Amplitude(volt)');
xlabel('time(sec)');
title('Encoded signal at Duobinary-Encoder');
subplot(3,1,3);
plot(t1,decoded digital code bit,'lineWidth',2.5);
grid on;
axis([ 0 bp*length(coded) -1.5 1.5]);
ylabel('Amplitude(volt)');
xlabel('time(sec)');
title('Digital Signal at Duobinary-Decoder');
%% With Error Plots
%orignal signal, staircase signal, recieved staircase signal, rec_low_pass
figure('Name','Adaptive Delta Modulation with noise','NumberTitle','off');
plot(t,x,'DisplayName','Message signal');
title('Adaptive Delta Modulation with error');
xlabel('Time (in sec)');
ylabel('Amplitude (in volts)');
hold 'on';
stairs(t,xq,'DisplayName','Staircase signal with noise');
```

```
stairs(t,rec_staircase_error,'DisplayName','Recieved Staircase signal with noise');
plot(t,y error, 'DisplayName', 'Recieved signal with noise');
hold 'off';
legend
%Binary signals
bp = .00001; %bit period
br = 1/bp; %bit rate
digital code error bit = [];
for n=1:1:length(digital code error)
    if digital code error(n) == 1
      se=ones(1,100);
    else
        se=-1*ones(1,100);
    end
     digital code error bit=[digital code error bit, se];
end
coded error bit = [];
for n=1:1:length(coded error)
    if coded error(n) == 2
       c se=2*ones(1,100);
    elseif coded error(n) == -2
        c_{se} = -2 \times (1,100);
    else
        c_{se} = zeros(1,100);
    end
     coded error bit=[coded error bit c se];
end
decoded_digital_code_error_bit = [];
for n=1:1:length(decoded digital code error)
    if decoded digital code error(n) == 1
       se=ones(1,100);
    else
        se=-1*ones(1,100);
    end
     decoded_digital_code_error_bit=[decoded_digital_code_error_bit se];
end
t1= bp/100:bp/100:100*length(x)*(bp/100);
figure('Name','Bit Transmission with error','NumberTitle','off')
subplot(3,1,1);
plot(t1,digital code error bit, 'lineWidth', 2.5);
grid on;
axis([ 0 bp*length(digital code error) -1.5 1.5]);
ylabel('Amplitude(volt)');
xlabel('time(sec)');
title('Digital Signal at Transmitter');
subplot(3,1,2);
plot(t1,coded error bit,'lineWidth',2.5);
axis([ 0 bp*length(coded error) -2.5 2.5]);
ylabel('Amplitude(volt)');
xlabel('time(sec)');
title('Encoded signal at Duobinary-Encoder');
subplot(3,1,3);
plot(t1, decoded digital code error bit, 'lineWidth', 2.5);
grid on;
axis([ 0 bp*length(decoded_digital_code) -1.5 1.5]);
ylabel('Amplitude(volt)');
```

```
xlabel('time(sec)');
title('Decoded signal at Duobinary-Decoder');
%% Parameters Calculations
disp(' ');
disp('Parameters')
disp(' ');
                                                                             %error
error = xor(digital code error, digital code);
disp('The error due to noise is');
disp(error);
Bit error rate = (sum(error(error==1)))/length(error);
disp('Bit Error Rate:');
disp(Bit_error_rate);
%SNR
my snr i = snr(xq, x error-xq);
my_snr_o = snr(y,y_error-y);
disp('Input SNR');
disp(my_snr_i);
disp('Output SNR');
disp(my snr o);
```

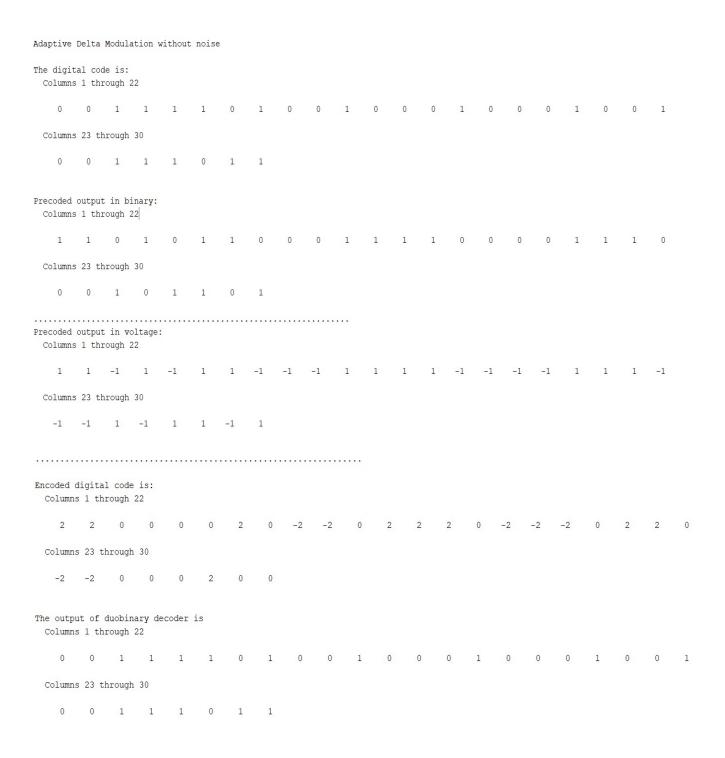
Communication_system_duobinary_error.m

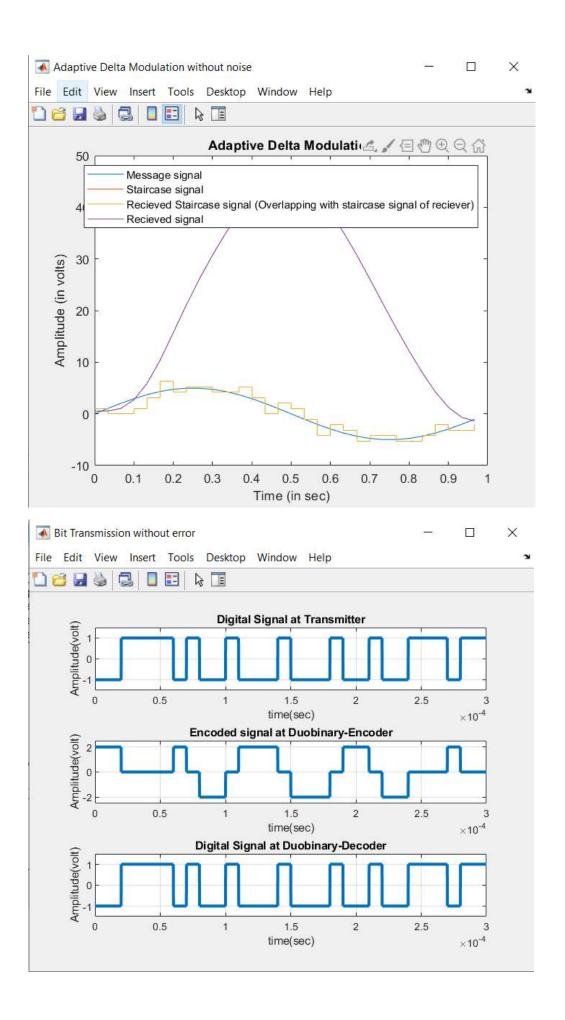
```
Am = 1;
             %amplitude
                                   %input('Enter the value for amplitude');
fm = 1;
             %frequency
                                   %input('Enter the value for frequency');
fs = 15*fm; %sampling freq
                                   %input('What is the sampling frequency?');
t = 0:1/fs:1;
%input signal
x = Am*sin(2*pi*fm*t);
delta = (2*pi*fm*Am)/fs;
[digital code, xq, zero track] = ad deltamod(x, delta);
%Transmittor
disp('The digital code is:');
disp(digital code);
disp(' ');
coded = Duobinary Encoder(digital code);
%duobinary_encoder
disp(' ');
disp('Encoded digital code is:');
disp(coded);
disp(' ');
coded = awgn(coded, -20);
decoded digital code = Duobinary Decoder(coded);
%duobinary decoder
disp('The output of duobinary decoder is');
disp(decoded digital code);
disp(' ');
[rec_staircase,my_signal] = ad_deltademod(decoded_digital_code,zero_track,delta);
%Reciever
```

```
y = lowpass(my_signal,1,fs);
error = xor(decoded digital code, digital code);
disp(' ');
disp('error');
disp(error);
figure('Name','Adaptive Delta Modulation(Noise at Duo-binary
Encoder)','NumberTitle','off');
plot(t,x,'DisplayName','Message signal');
title('Adaptive Delta Modulation');
xlabel('Time (in sec)');
ylabel('Amplitude (in volts)');
hold 'on';
stairs(t,xq,'DisplayName','Staircase signal');
stairs(t,rec_staircase,'DisplayName','Recieved Staircase signal at reciever');
plot(t,y,'DisplayName','Recieved signal');
hold 'off';
legend
```

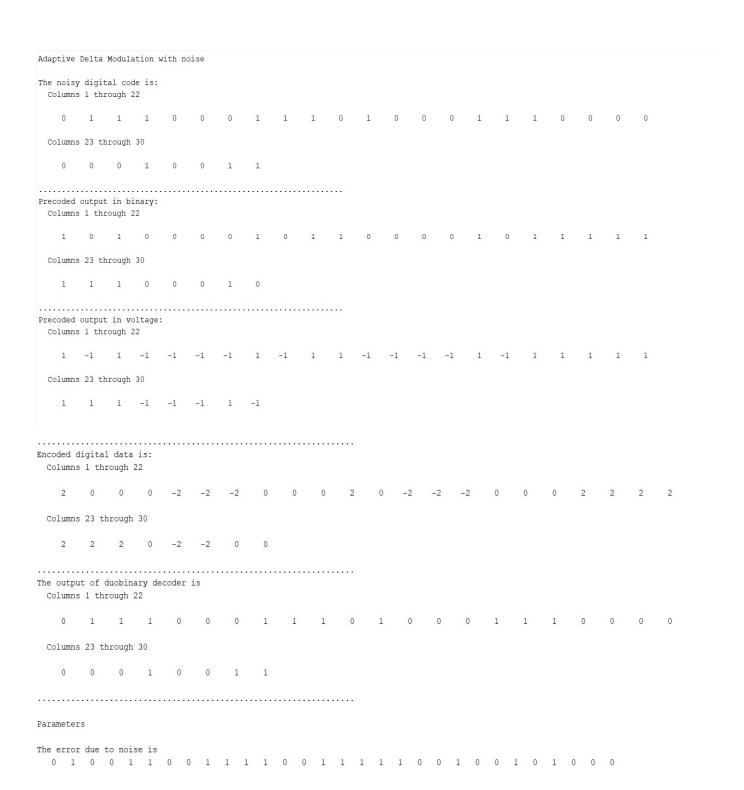
6. Output:

Without Noise Output and Plots:

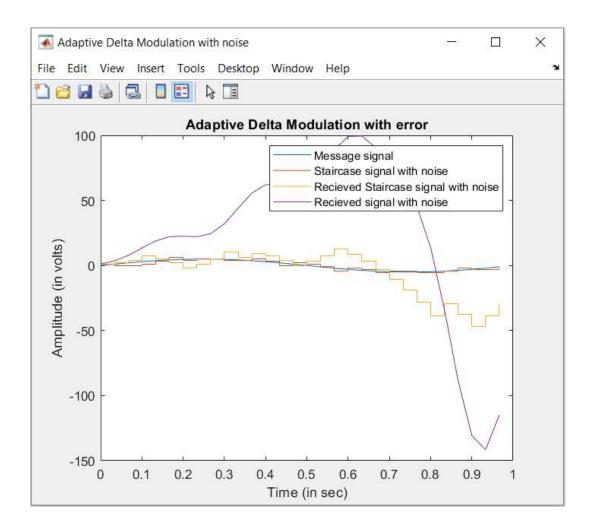


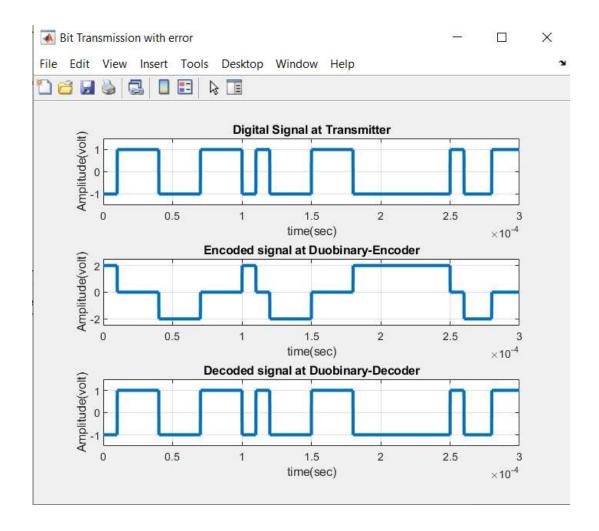


With Noise Output and Plots:



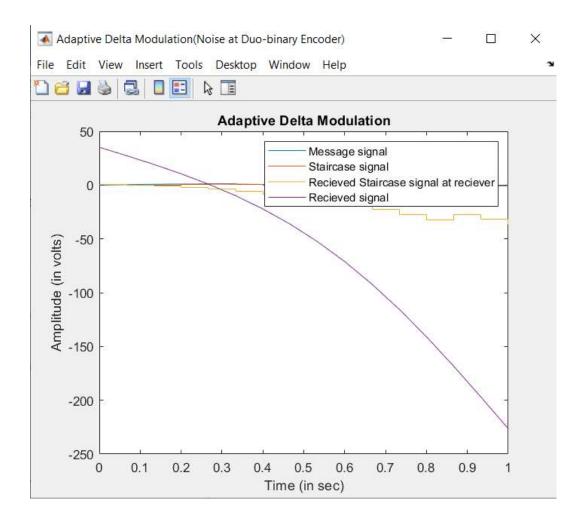
Bit Error Rate: 0.5000 Input SNR 1.3569 Output SNR -5.7311





When the error was introduced in between Duobinary Encoder:

The digit	cal co														
0	0	1	1	1	0	0	1	0	0	0	0	1	1	0	1
Precoded	outpu	t in bi	inary:												
1	1	0	1	0	0	0	1	1	1	1	1	0	1	1	0
Precoded	011+711														
					-1	-1	1	1	1	1	1	-1	1	1	-1
Encoded of	ligita	1 code	ie.												
2	2	0		0	-2	-2	0	2	2	2	2	0	0	2	0
The outpu	ı+ of .	du ob i n -	anır da	andon	ia										
0		0				0	0	0	0	0	0	0	1	0	0
error															
0 0	1	1 1	0	0 1	. 0	0 0	0	1 0	0	1					



7. Conclusion

An end-to-end Digital Communication System in form of Adaptive delta Modulation with duo-binary Signaling was observed with and without noise and the parameters such as error bit stream, bit-error rate and Signal to Noise Ratio(SNR) were calculated.

8. Bibliography

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