

# 1 Objectives

The objective for this lab was to build a quantizer for PCM and creating a Delta modulator in MATLAB. The secondary objective is to understand how to quantify the quantization noise of a PCM signal and the granulation noise of a delta modulated signal.

# 2 Procedure

The MATLAB code that was used in the lab is found in Appendix A.

We began defining a signal that will be Pulse Code Modulated and Delta Modulated. The signal that was used in the lab was defined as:

$$y(t) = \sin(2\pi t) + \sin(4\pi t) + \sin(5\pi t) + \sin(9\pi t) + \sin(10\pi t) + \sin(24\pi t)$$

The time interval was from 0 to 1 with a time step of 0.001

The signal was quantized using the custom function `quantizedSample`. The variable `quantizedData` holds integer values that represent the level of the signal. These values would normally be encoded into a binary stream and then transmitted but for the purpose of this lab that wouldn't be necessary. The resulting data was reconstructed using the custom function `reconstructedData`.

# 3 Results

## A MATLAB Code

```
t = 0:0.001:1;
fs = 1000;
L = 4;
signal = sin(t* 2 * pi) + sin(t* 4 * pi) + sin(t* 5 * pi)...
        + sin(t* 9 * pi) + sin(t* 10 * pi) + sin(t* 24 * pi);
signal = signal/max(abs(signal));
quantizedData = quantizedSample(signal, L);

reconstructedData = reconstructQuantized(max(abs(signal)),...
    L, quantizedData);

plot(t, signal, t, reconstructedData);

q = signal - reconstructedData;
theoreticalError = max(abs(signal)).^2/(3*L^2)
measuredError = mean(q.^2)
var(q)
plot(t, q);

fs= 100*10;
t = -3:1/fs:3;
signal = exp(-2*t.^2).*cos(pi*t);

sampleRatio = 4;
```

```

optimalStepSize = pi * max(myDerivative(signal,1/fs))*max(signal)/fs;

[delSignal,estime] = delQuantization(optimalStepSize, signal, fs,...
                                     fs/sampleRatio);
recon1 = cumulativeSum(delSignal);
resampled = upsample(delSignal, sampleRatio);

resampled = cumulativeSum(resampled * optimalStepSize);

test1 = cumulativeSum(delSignal * optimalStepSize);
test2 = cumulativeSum(delSignal) * optimalStepSize;
resampled = resampled(1:length(t));
plot(t, signal, 'b.', t - 5/fs, resampled, 'g');

filtered = filter(DelFilter(fs),[1],resampled);
plot(t, signal, 'b.', t, filtered, 'g');

shift1 = 5;
shift2 = 0;
amp = 1%/0.14;
granError1 = signal(1+shift1:end) - resampled(1:end-shift1);
granError2 = signal(1+shift2:end) - amp*filtered(1:end-shift2);
plot(t(1+shift1:end), granError1, t(1+shift2:end), granError2);

```

## B Supplementary Code

### B.1 quantizedSample

```

function data = quantizedSample(data, numSamples)
maxValue = max(data)*2;
data = data * numSamples / maxValue;
data = floor(data);
end

```

### B.2 reconstructQuantized

```

function data = reconstructQuantized(maxValue, numbLevels, data)
levelValue = maxValue * 2 / numbLevels;
data = data*levelValue + levelValue / 2;
end

```

### B.3 myDerivative

```

function dataout = myDerivative(dataIn, samplePeriod);
dataout = (dataIn(2:end) - dataIn(1:end-1))/samplePeriod;
end

```

### B.4 delQuantization

```

function [data,value] = delQuantization(stepSize, signal,...
                                     signalSampleRate, newSampleRate)

```

```

    signal = downsample(signal, ceil(signalSampleRate../newSampleRate));
    data = zeros(1,length(signal));
    error = zeros(1,length(signal));
    value = zeros(1,length(signal));
    data(1) = 1;
    value(1) = stepSize;
    for i = 2:length(signal)
        error(i) = signal(i-1) - value(i-1);
        if(error(i) > 0)
            data(i) = 1;
        else
            data(i) = -1;
        end
        value(i) = value(i-1) + data(i) * stepSize;
    end
end

```

## B.5 cumulativeSum

```

function dataOut = cumulativeSum(dataIn)
    dataOut = zeros(1,length(dataIn));
    dataOut(1) = dataIn(1);
    for i = 2:length(dataIn)
        dataOut(i) = dataOut(i-1) + dataIn(i);
    end
end

```

## B.6 DelFilter

```

function H = DelFilter(Fs)
%DELFILTER Returns a discrete-time filter object.

% MATLAB Code
% Generated by MATLAB(R) 9.4 and Signal Processing Toolbox 8.0.
% Generated on: 06-Mar-2019 20:53:25

% Equiripple Lowpass filter designed using the FIRPM function.

% All frequency values are in Hz.
%Fs = 100; % Sampling Frequency

Fpass = 1;           % Passband Frequency
Fstop = 4;           % Stopband Frequency
Dpass = 0.057501127785; % Passband Ripple
Dstop = 0.0001;      % Stopband Attenuation
dens = 20;           % Density Factor

% Calculate the order from the parameters using FIRPMORD.
[N, Fo, Ao, W] = firpmord([Fpass, Fstop]/(Fs/2), [1 0], [Dpass, Dstop]);

% Calculate the coefficients using the FIRPM function.

```

```
b = firpm(N, Fo, Ao, W, {dens});  
Hd = dfilt.dffir(b);  
H = Hd.Numerator;  
% [EOF]
```