## SS EXPERIMENT LAB 3

**TITLE**: Perform decimation and calculate energy and power of a signal

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**OBSERVATION**: In this lab, I learned how to decimate and calculate energy and power using MATLAB.

1. The operation of signal dilation (or decimation or down-sampling) is defined by

$$y(n) = x(nM)$$

where the sequence x(n) is down-sampled by an integer factor M. For example, if

$$x(n) = \{\dots, -2, 4, 3, -6, 5, -1, 8, \dots\}$$

then the down-sampled sequences by a factor 2 are given by

$$y(n) = \{\dots, -2, 3, 5, 8, \dots\}$$

Develop a MATLAB function dnsample that has the form to implement the above operation.

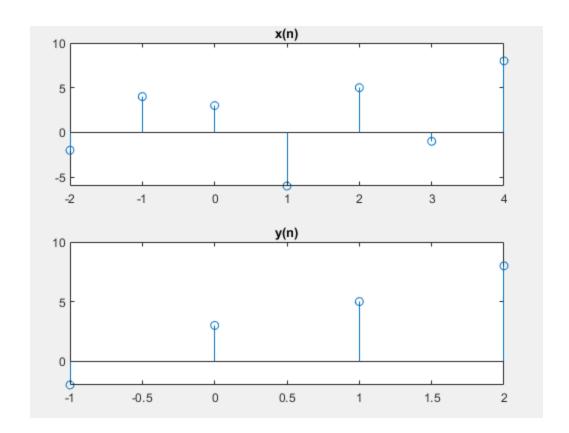
```
function [y,m] = dnsample(x,n,M)
% Downsample sequence x(n) by a factor M to obtain y(m)
```

```
n = -2:1:4;
x = [-2,4,3,-6,5,-1,8];
[y,m]= dnsample(x,n,2);

subplot(2,1,1);
stem(n,x);
title('x(n)');

subplot(2,1,2);
stem(m,y);
title('y(n)');

function [y,m]=dnsample(x,n,M)
param=n/M;
samp=fix(param)==param;
y=x(samp==1);
m=n(samp==1)/M;
end
```



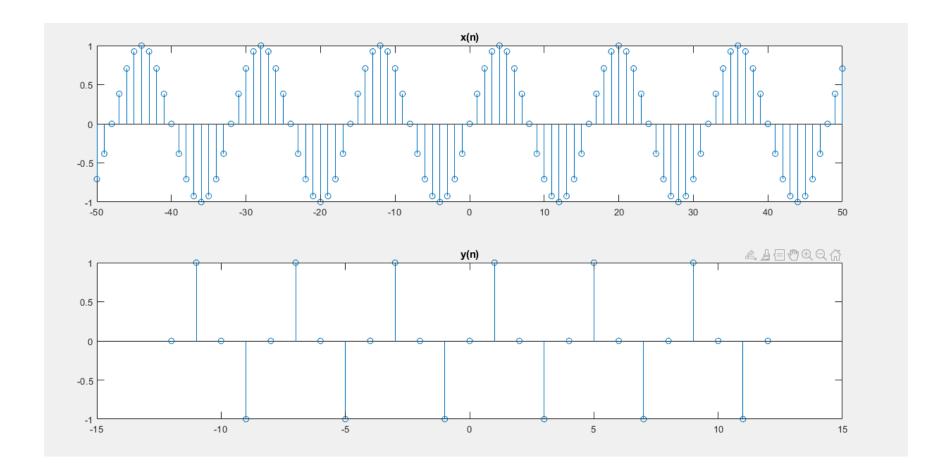
2. Generate  $x(n) = \sin(0.125\pi n)$ ,  $-50 \le n \le 50$ . Decimate x(n) by a factor of 4 to generate y(n). Plot both x(n) and y(n) using subplot, and comment on the effect of down-sampling on signal.

```
n = -50:1:50;
x = sin(0.125*pi*n);
[y,m] = dnsample(x,n,4);

subplot(2,1,1);
stem(n,x);
title('x(n)');

subplot(2,1,2);
stem(m,y);
title('y(n)');

function [y,m] = dnsample(x,n,M)
    param = n/M;
    samp = fix(param) == param;
    y = x(samp == 1);
    m = n (samp == 1)/M;
end
```



For a given speech signal, perform the following operations:

a) Read/load the voice file in MATLAB using in build 'audioread' function

```
[x, fs] = audioread('LDC93S1.wav');
% read audio file// x- speech sample, fs is the sampling frequency
```

- b) Write a MATLAB script (function) which segment the speech signal "20 ms" duration without overlapping
- c) Calculate the energy and power of the data available in each segment.
- Q3. d) Plot the *energy and power* of signal for each segment.

```
[x, fs] = audioread('LDC93S1.wav');
 segment = 0.02;
 inc = fs * segment;
 [E,p]=cal(x,inc);
 ll=[1:1:size(E)];
 12=[1:1:size(p)];
 subplot (3,1,1);
 plot(x);
 title('Amplitude');
 subplot (3,1,2);
 plot(E);
 title('Energy');
 subplot (3,1,3);
 plot(p);
 title('Power');
function [E,p]=cal(x,inc)
     E=[];
 p=[];
 sum=0;
 count=0;
for i=1:1:size(x)
     if(mod(i,inc)==0)
      E(end+1)=sum;
      p(end+1)=sum/count;
     sum=0;
     count=0;
     else
     sum=x(i)*x(i)+sum;
     count=count+1;
      end
 end
 if (mod(size(x),inc)~=0)
      E(end+1)=sum;
      p(end+1)=sum/count;
 end
 end
```

