

# AHSANULLAH UNIVERSITY OF SCIENCE AND TECHNOLOGY Department of Electrical and Electronic Engineering

#### **Project**

Course Name: Digital Signal Processing Lab

## Submitted by

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### Task-1

Program to implement a sampling rate conversion.

## <u>(i)</u>

```
Let,
Up-sampling factor = 3

Down-sampling factor = 2

Input frequency, f1 = 100 Hz & f2 = 200 Hz
```

# <u>(ii)</u>

```
x [n]=sin(2*pi*f1*n)+sin(2*pi*f2*n)
= sin(2*pi*100*n) + sin(2*pi*200*n)
x = sin(2*pi*100)+sin(2*pi*200) [let n =1]
```

#### (iii) Code:

```
clc;
clear all;
close all;
hsin = dsp.SineWave(1,[100 200], 'SampleRate',
5000, 'SamplesPerFrame', 50);
% Create a FIR rate converter filter.
% The default interpolation factor is 3 and
decimation factor is 2.
hfirrc = dsp.FIRRateConverter;
input = step(hsin);
output = step(hfirrc, input);
% Plot the original and resampled signals.
ndelay=round(length(hfirrc.Numerator)/2/hfirrc.D
ecimationFactor);
indx = ndelay+1:length(output);
x = (0:length(indx) -
1) /hsin.SampleRate*hfirrc.DecimationFactor/hfirr
c. Interpolation Factor;
stem((0:38)/hsin.SampleRate, input(1:39));
hold on;
stem(x,
hfirrc.InterpolationFactor*output(indx), 'r');
legend('Original','Resampled');
```

# (iv) Output:



