

Presentation on

Digital Signal Processing Assignment

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Matriculation no. 300 428 36

Task 1

Task 1 (a)

$$H(z) = 1/2 + (1/3) \frac{z}{z-1} - \frac{1}{2z-1}$$

Task 1 (b)

$$H(z) = 4 + \frac{1}{2z-1} - \frac{1}{2z}$$

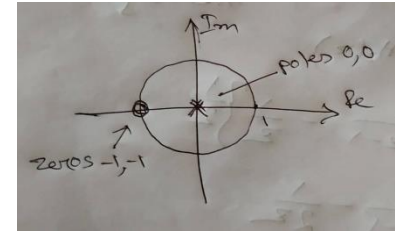
Task 1 (c)

a) $h(n) = 1/2 \delta(n) + \delta(n-1) + 1/2 \delta(n-2)$

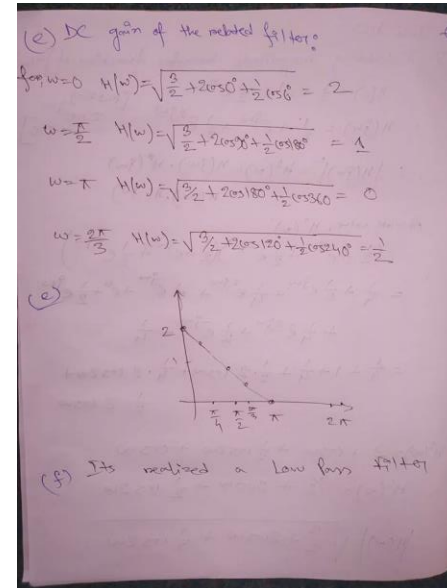
b) $H(z) = 1/2 + 1/z + 1/2z^2$

Zeros, at -1, -1

Poles at 0.0



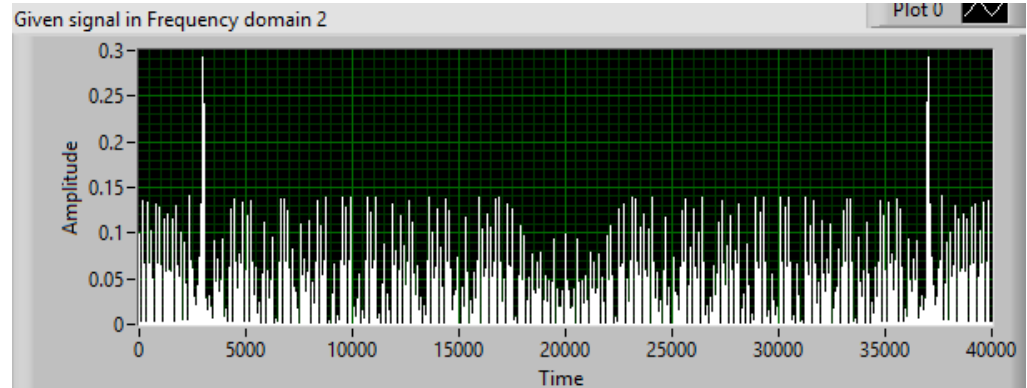
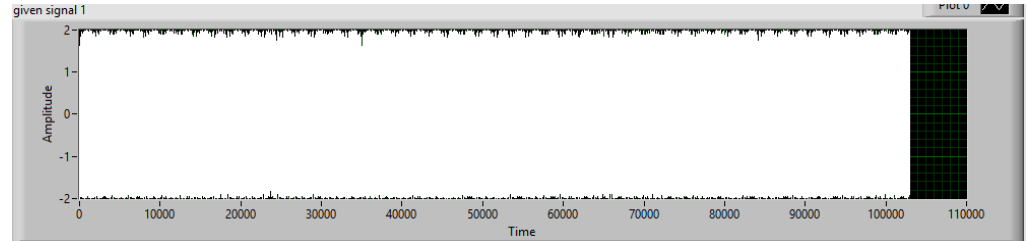
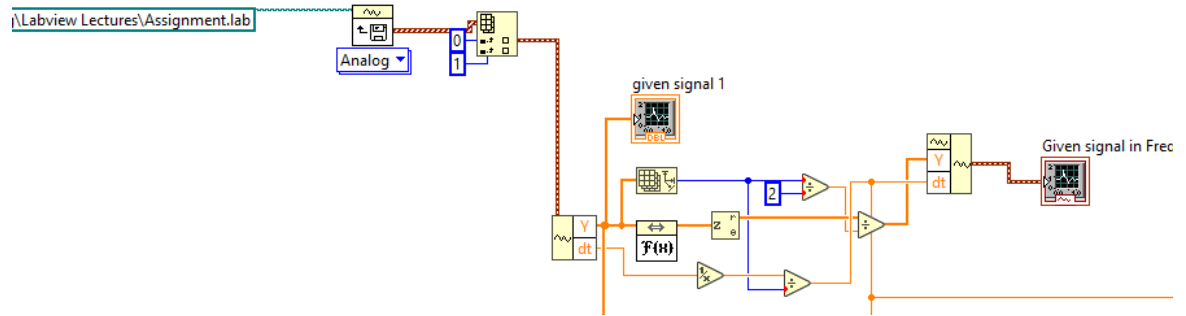
d) $|H(z)| = \sqrt{1.5 + 2\cos w + 0.5\cos 2w}$



Task 2

The sample frequency of the signal is 40KHz and the signal is modulated using a 3KHz carrier signal.

The message signal is a binary signal, just two dominant phases are available in the BPSK modulated signal. The 180° shift of phase codes a '1' and the 0° of phase codes a '0'.



Task 2

Designing Band pass filter

Zeros

For zero at +1 \rightarrow highpass

For zero at -1 \rightarrow lowpass

Zeros at 1, -1 \rightarrow bandpass

$$H(z) = \frac{(z-1)(z+1)}{(z-Re^{j\omega})(z-Re^{-j\omega})}$$

Poles

$F1=3400\text{Hz}$

$F2=2700\text{Hz}$

$BW=f2-f1=3400-$

$2700=700\text{Hz}$

$Fs=40\,000\text{Hz}$

$R=1-(BW/fs)*\pi$

$=1-(700/40\,000)*\pi$

$R=0.945$

$\omega = 2\pi(f_0/Fs)$

$f_0=3000\text{Hz}$

$\omega = 0.15\pi = 27^\circ$

$P1=Re^{j\omega}$

$P2=Re^{-j\omega}$

$P1=0.945(\cos 27^\circ + j\sin 27^\circ)$

$P2=0.945(\cos 27^\circ - j\sin 27^\circ)$

The denominator of the

Transfer function

$(z-P1)(z-P2)$

$= (z-0.945e^{j27})(z-0.945e^{-j27})$

$= z^2 - 1.684z + 0.893$

$$H(z) = \frac{z^2 - 1}{z^2 - 1.684z + 0.893}$$

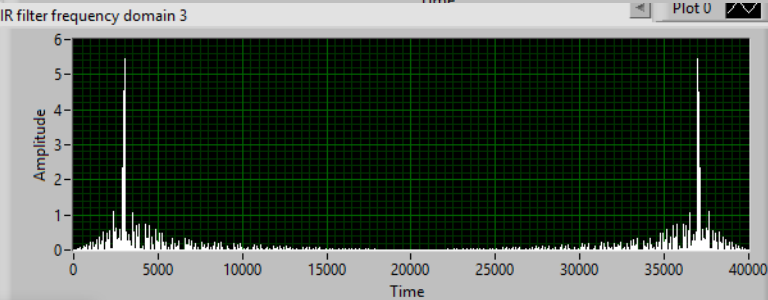
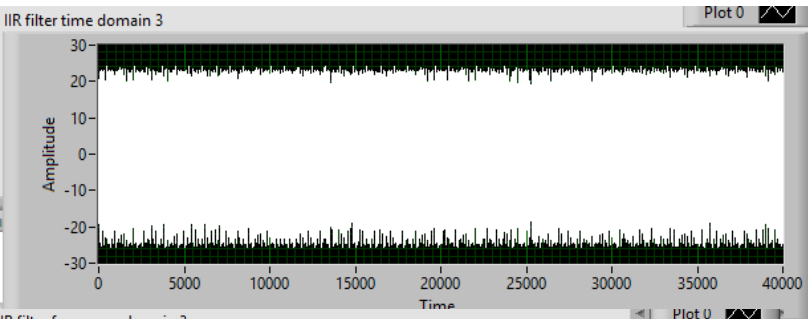
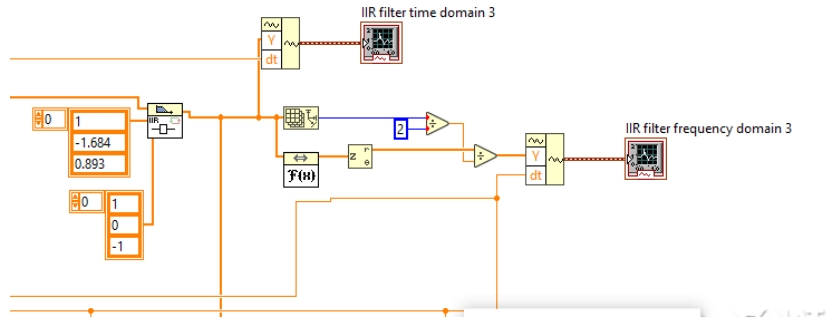
Reverse coefficients $a0=1$

$a1=-1.684$ $a2=0.893$

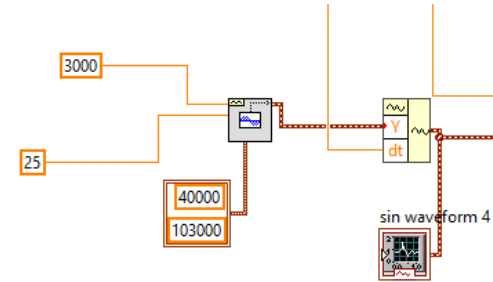
Forward coefficients

$b0=1$ $b1=0$ $b2=-1$

Implementing IIR Filter



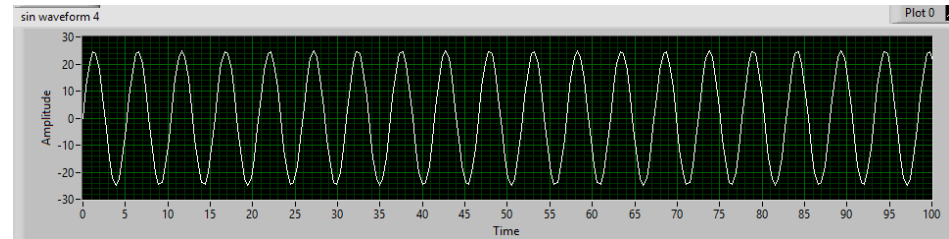
Generating a carrier clock using a sine wave



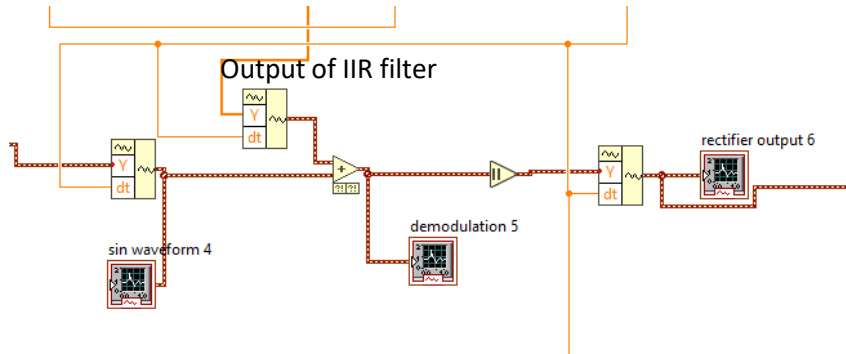
F_s is the sampling rate in samples per second 40000Hz

#s is the number of samples in the waveform 103000 (from Source signal)

F_c=3000 Amplitude =25 (close to the IIR output)



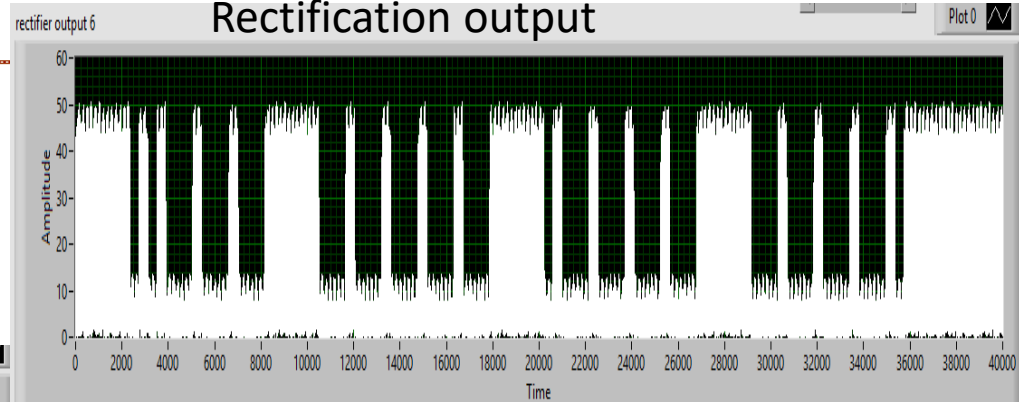
Demodulation



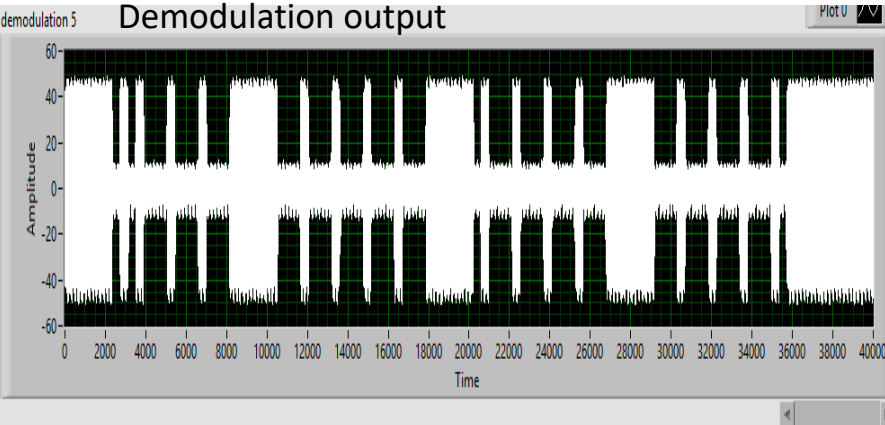
Rectification

For rectification Absolute value function
Is used

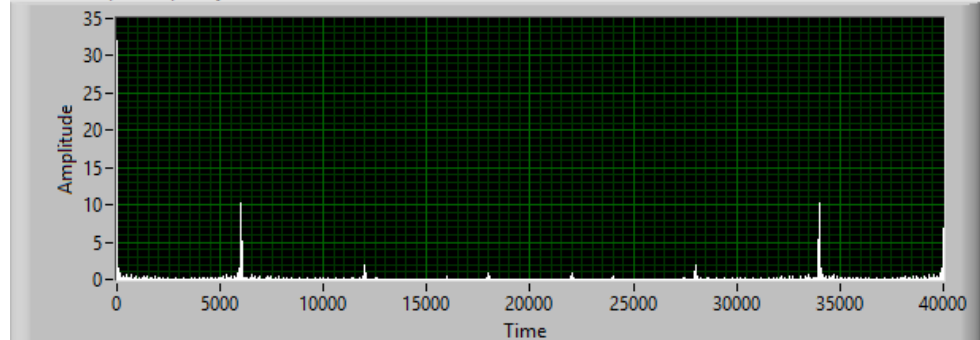
Rectification output



Demodulation output

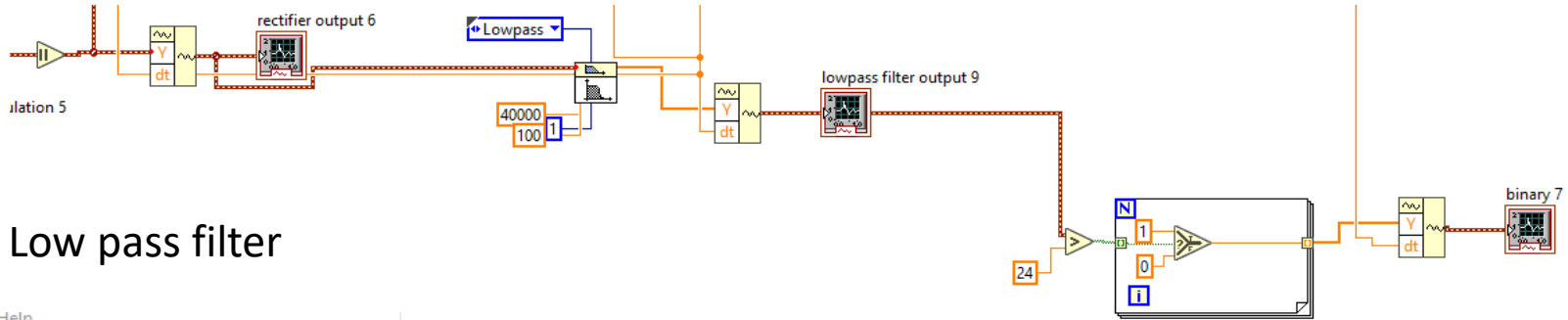


rectifier output frequency domain 4



Restricting high frequency components

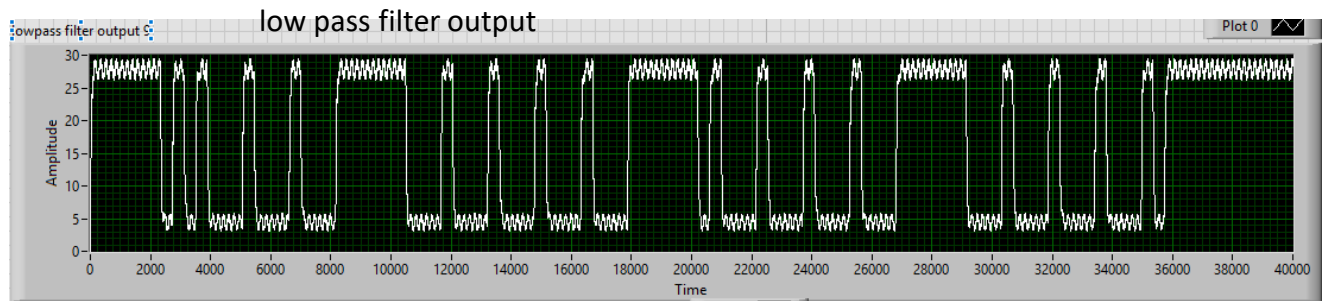
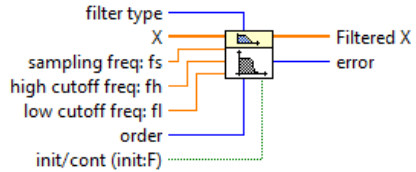
By using low pass filter



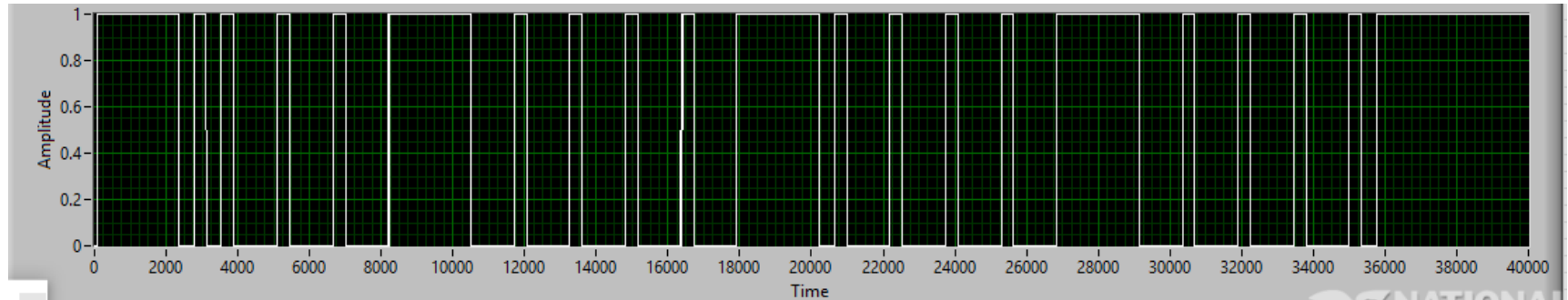
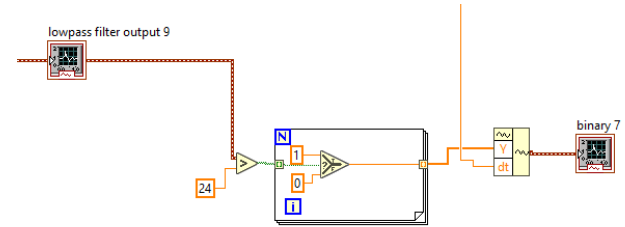
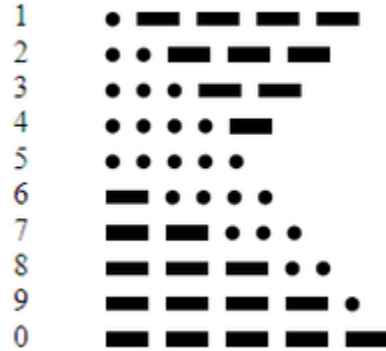
Low pass filter

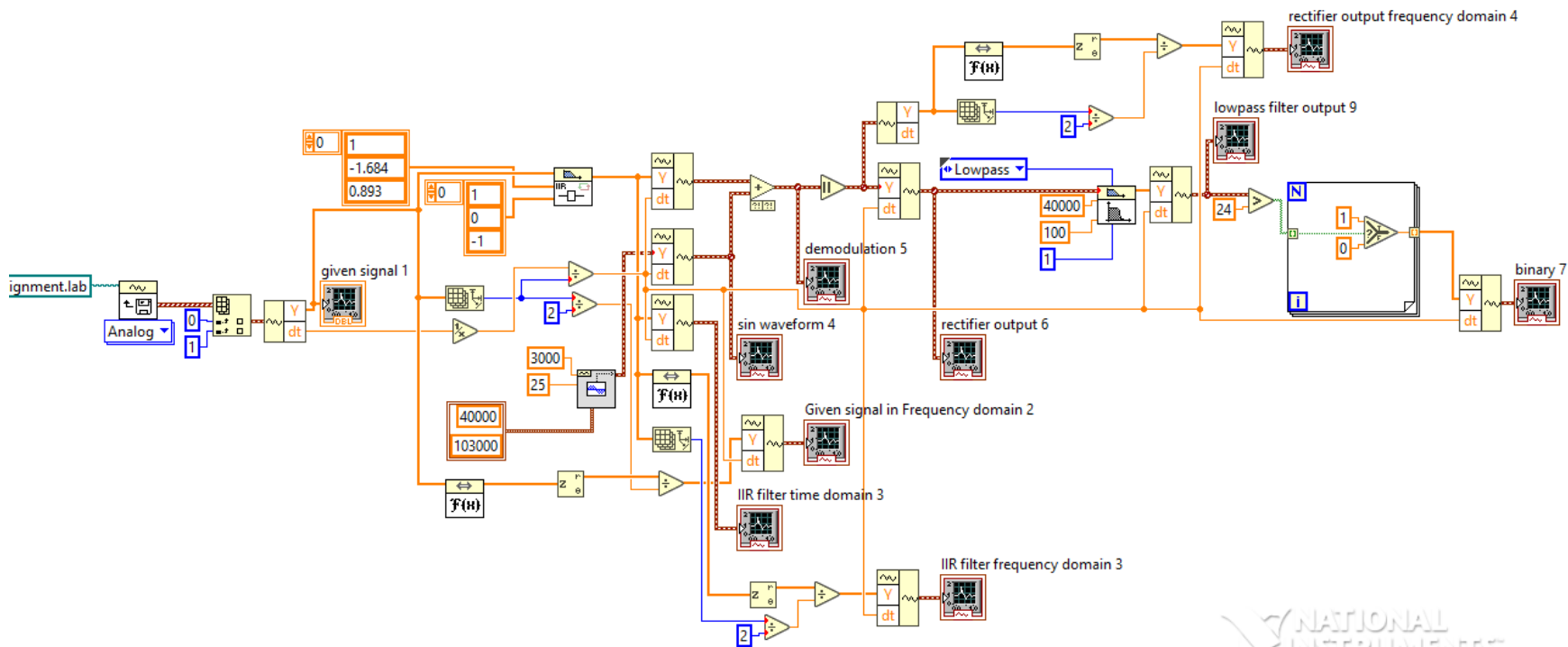
Context Help

NI_AALBase.lvlib:Butterworth Filter.vi



For understanding the signal , the output is converted in binary form





Thank you....