

# Samplingtheorem -- Overview

## Sampling Theorem



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## OBJECTIVES:

At the end of performing this experiment, learners would be able to:

- Describe the concept of sampling a time varying signal
- Obtain the naturally sampled signal from given input
- Understand the working of LF398 IC (sample-and-hold circuit)

## EQUIPMENT:

- IC LF398
- Signal generator
- Resistors – 47 k $\Omega$ , 1.5 k $\Omega$
- Capacitor – 0.1  $\mu$ F
- +/- 15V DC Power Supply
- Digital Storage Oscilloscope & probes
- Connecting wires & Bread Board

## DESIGN:

For First ,Second and Third order filter

$$f = 1\text{kHz}$$

$$R = ?$$

$$C = 0.1\mu\text{F}$$

$$f = \frac{1}{2\pi RC}$$

$$R = 1.59\text{k}\Omega \text{ taken } 1.5\text{k}\Omega$$

## THEORY:

- LF398 is a monolithic sample-and-hold circuit utilizing BI-FET technology for accurate fast acquisition of input signal.
- A sample and hold circuit is an analog device that samples (captures) the voltage of a continuously varying analog signal and holds (locks) its value at a constant level for a specified minimum

period of time (hold time). They are typically used in analog-to-digital converters to eliminate variations in input signal that can corrupt the conversion process.

- A band limited signal of finite energy which has no frequency components higher than  $W$  Hz is completely described by specifying the values of the signal at instants of time separated by  $1/2W$  seconds

#### Reference reading:

B Kanmani, "Some applications of the combination: LM-741 and LF - 398", WASET CESSE 2009: International conference on Computer, Electrical and Systems science and Engineering, Rome, 28th-30th April, Italy, 2009. Volume 52, April 2009, ISSN: 2070-3724, pages 335-340.

#### Acknowledgement

Mr. Shreenivas B for converting laboratory experiment to Tektronix courseware format

## Sampling theorem -- Procedures

### Step 1

#### Circuit setup:

Build the following circuit with given component values

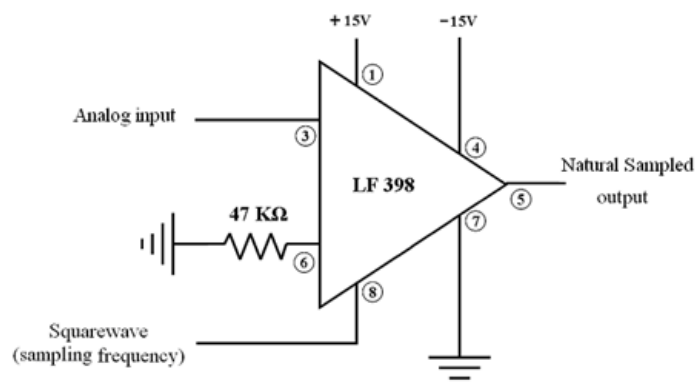


Figure: The circuit used to generate 'Natural-sampled' waveform

### Step 2

- Use a signal generator to generate analog input and sampling (square wave signal). The analog input will be set to 1 kHz Sine wave (or triangular wave) and sampling signal will be 15-20 kHz Square-wave of 20% duty cycle.

- Turn on the supply of the circuit and enable signal generator that is feeding signal to the circuit.

### Step 3

- Connect the DSO probe – CH1 at analog input (pin # 3 of LF398 IC), CH2 at sampling signal input (pin # 8 of LF398 IC) and CH3 at output (pin # 5 of the LF398 IC).
- Perform Autoset on DSO and capture the output signal.

### Step 4

- Configure PEAK-to-PEAK measurement on the input and output signal
- Observe the output Message with dc-offset and sampled waveform

### Step 5

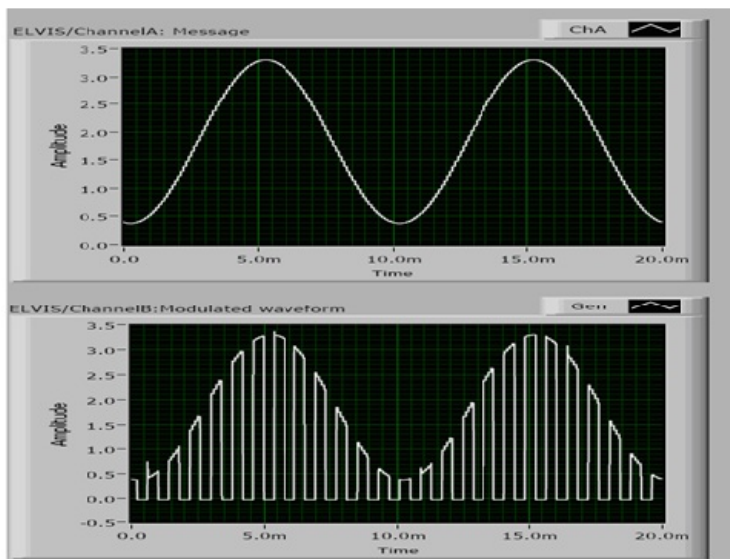


Figure: The sinusoidal message with dc-offset and the corresponding sampled waveform

### Step 6

- Observe the output Message without dc-offset and sampled waveform

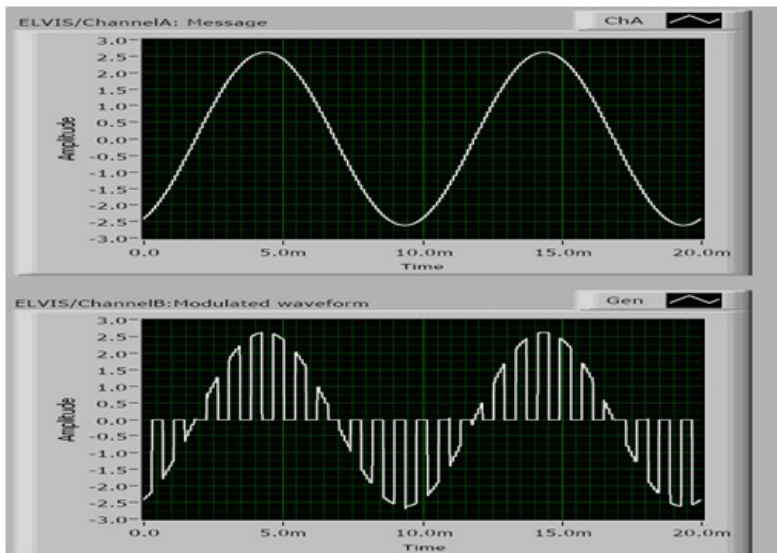
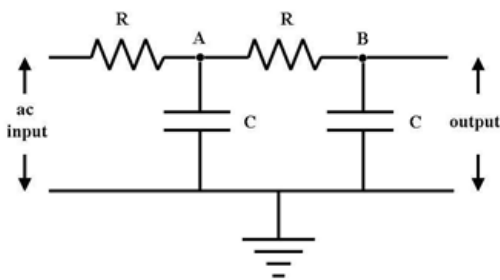


Figure: The sinusoidal message without dc-offset and the corresponding sampled waveform

## Step 7

- Record the measurement and Observe – input, output on DSO  
The output of the LF398 is given to an Low Pass Filter ( First or Second or Third order)  
As shown in the following figure



## Step 8

### Observations:

Obtain the minimum sampling frequency for 1KHz message signal when recovery is using

- first order LPF is \_\_\_\_\_
- second order LPF is \_\_\_\_\_
- third order LPF is \_\_\_\_\_

## Step 9

- Open-ended Question / Can you answer this?

What will be the effect on output waveform if:

- 1) Order of the LPF is increased

