



Lab(3)  
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Design of Measurement Systems, Sp. 2015

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# LAB Goals:

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## Understanding:

- While and For Loops
- For Loop and 2D Arrays
- Shift Registers
- Case Structure
- Formula Node
- ▶ Demonstrating different types of signals.
- ▶ Understanding FFT through various signal examples.
- ▶ Understanding different types of Filters through several set of examples.



# Structures

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- While and For Loops
- For Loop and 2D Arrays
- Shift Registers
- Case Structure
- Formula Node

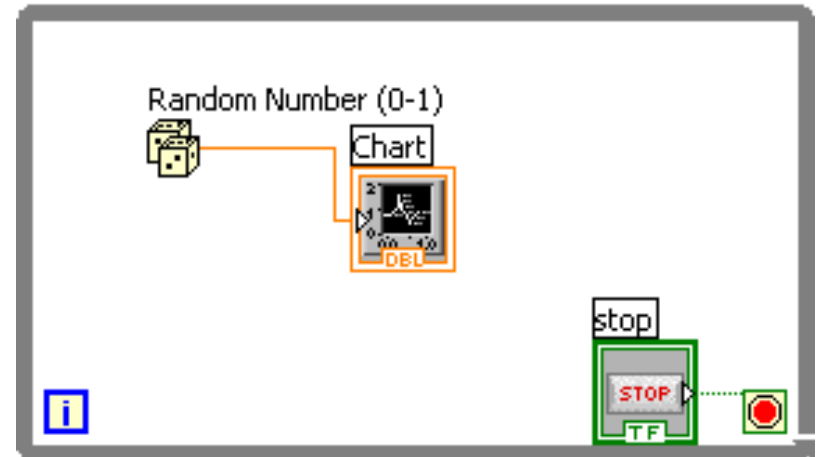


# Loops

## ▶ While Loops

- ▶ Have Iteration Terminal
- ▶ Always Run at least Once
- ▶ Run According to Conditional Terminal (Two different states)

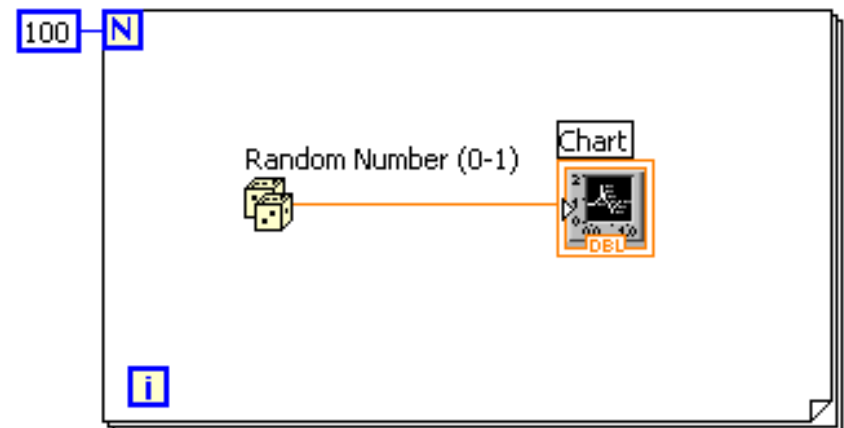
While Loop



## ▶ For Loops

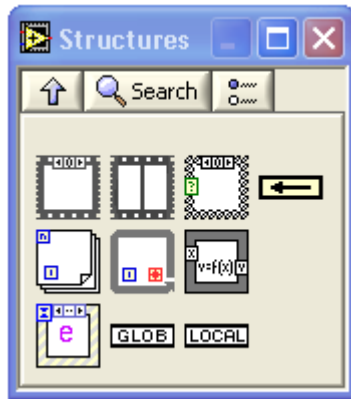
- ▶ Have Iteration Terminal
- ▶ Run According to input N of Count Terminal

For Loop

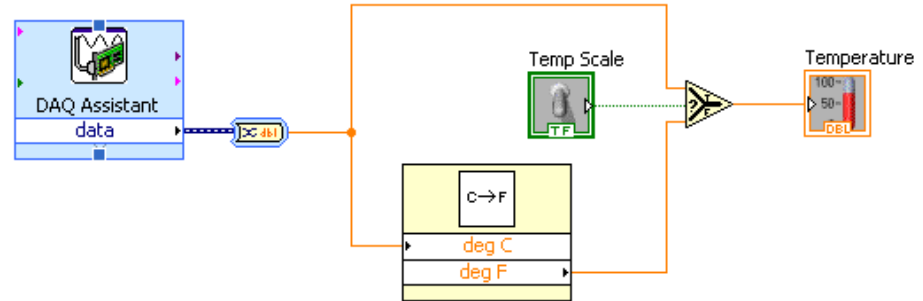


# Loops (cont.)

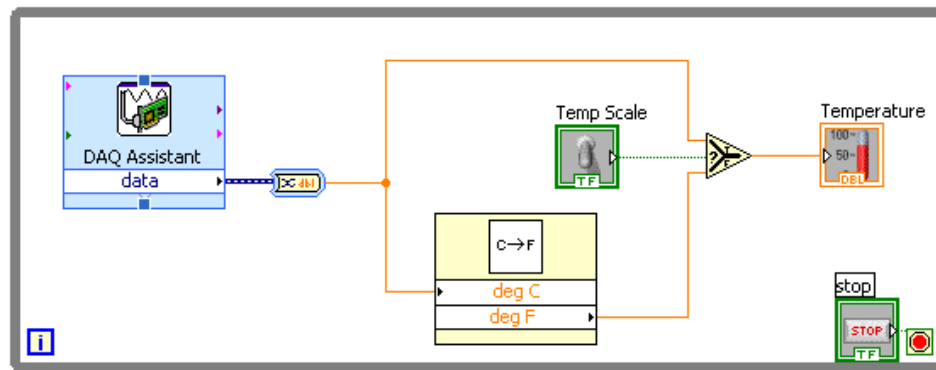
1. Select the loop



2. Enclose code to be repeated



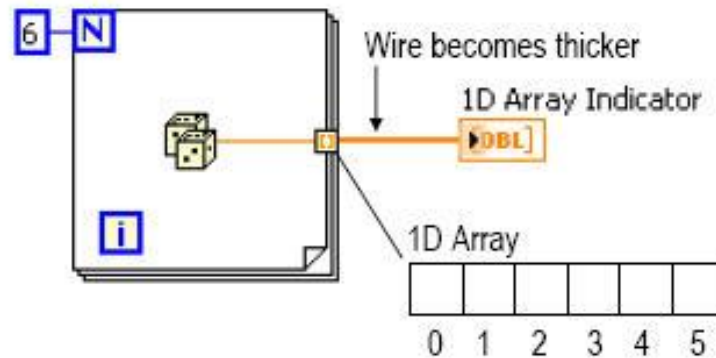
3. Drop or drag additional nodes and then wire



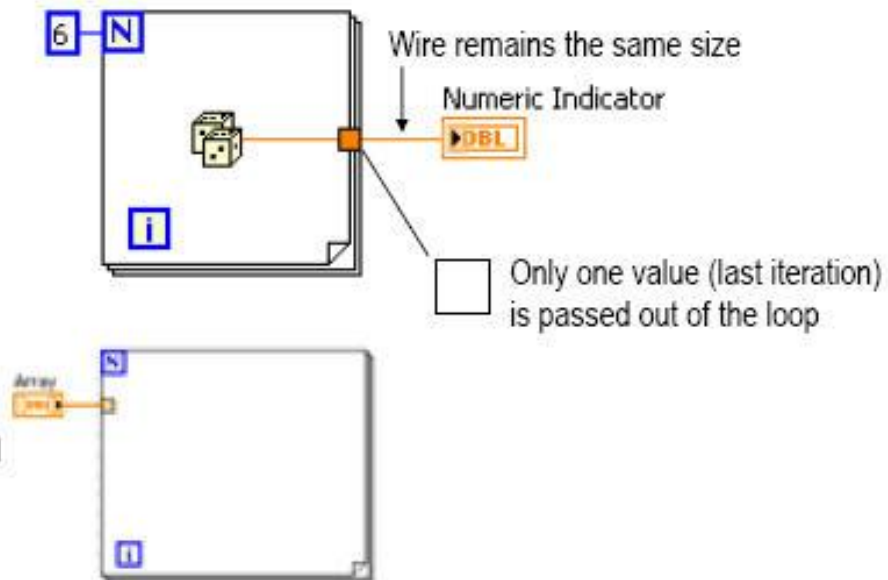
# Auto-Indexing

- Loops can accumulate arrays at their boundaries with auto-indexing
- For Loops auto-index by default
- While Loops output the final value by default
- Right-click on tunnel and enable/disable auto-indexing
- Auto-indexing on Loop input converts arrays to indexed elements (default), which can be disabled at the node.

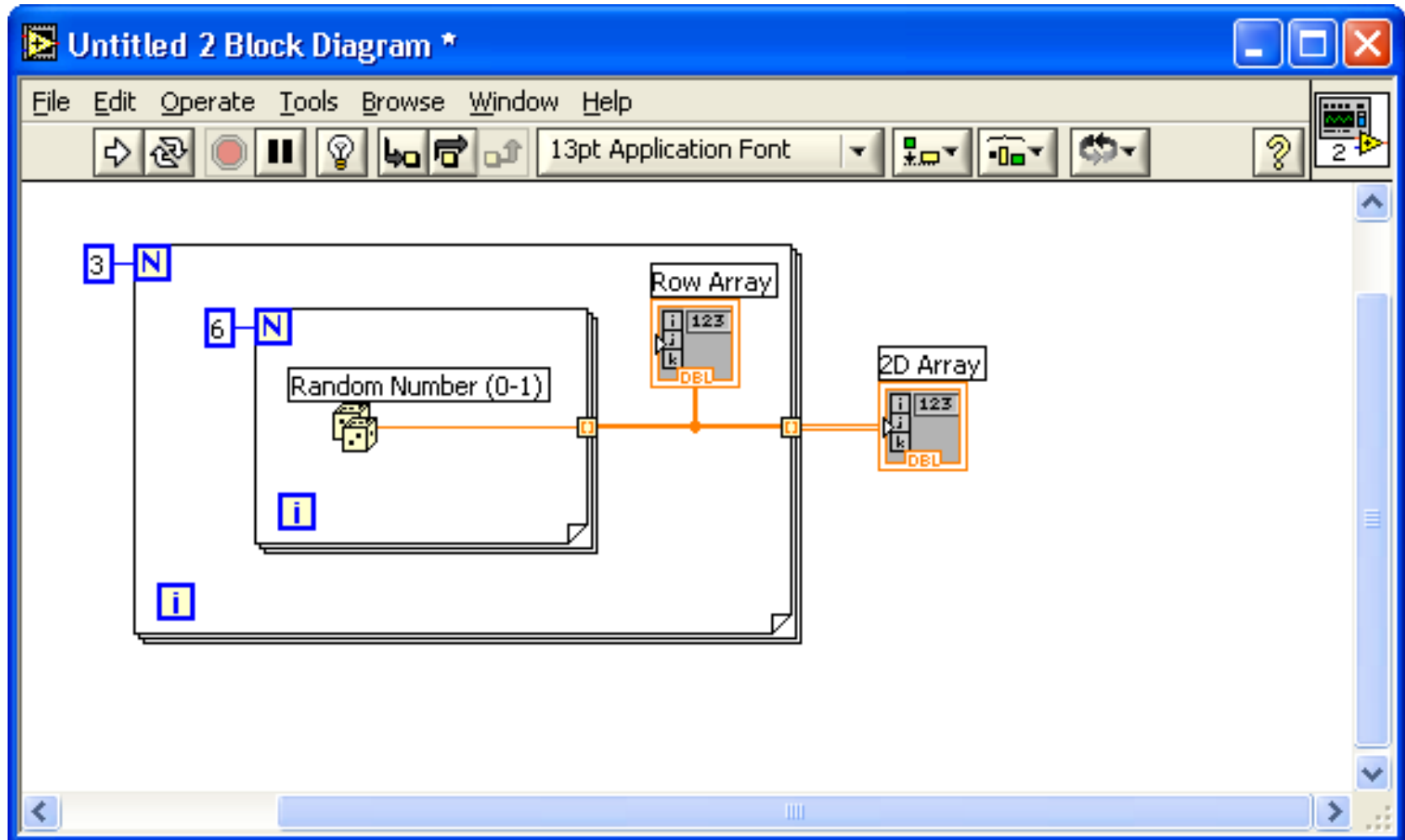
## Auto-Indexing Enabled



## Auto-Indexing Disabled

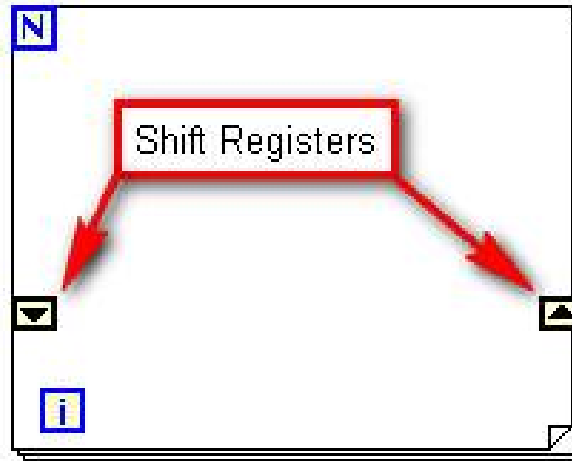


# Creating 2D Arrays with For Loops (Auto-Indexing)



# Shift Registers

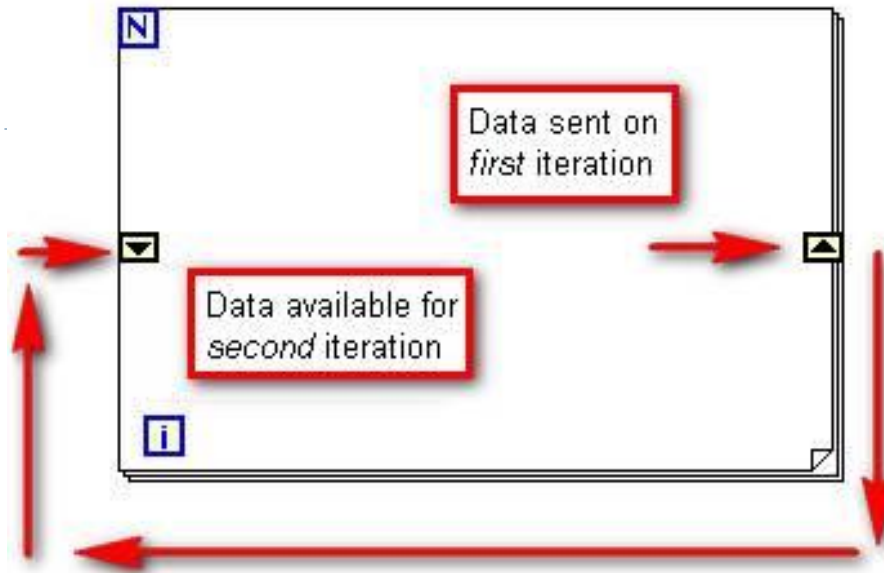
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- Selected via right-clicking the frame.
- Enables the result of an iteration to be passed to the next iteration.
- Can be used for any data type







- Data comes into the loop by way of the shift register on the left side of the for loop and is passed to the next iteration of the loop through the shift register on the right side of the for loop.
- The initial value is set by wiring to the left terminal and the final iterations value is output at the right terminal.



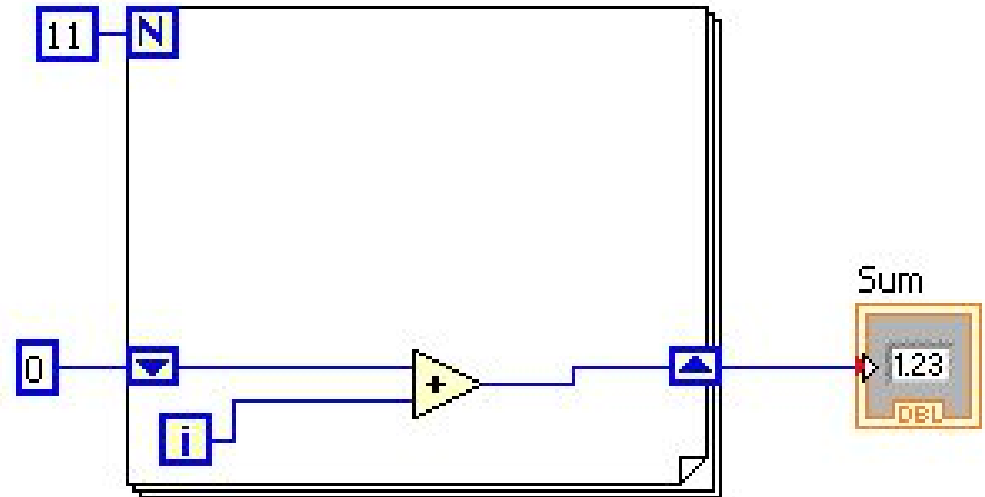
# Example

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## ▶ C/C++ Code:

```
int i, n, sum;  
n=11;  
sum=0;  
for (i=0; i<n; i++)  
{  
sum = sum + i;  
}
```

## LabVIEW VI:



# Exercise (1)

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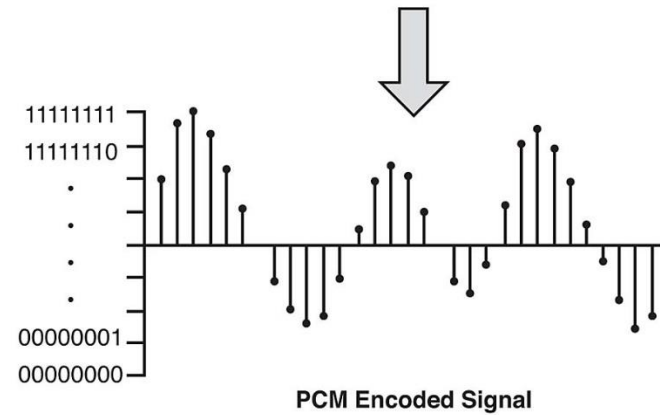
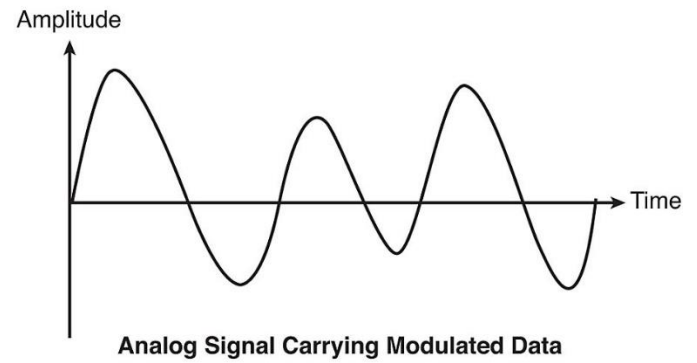
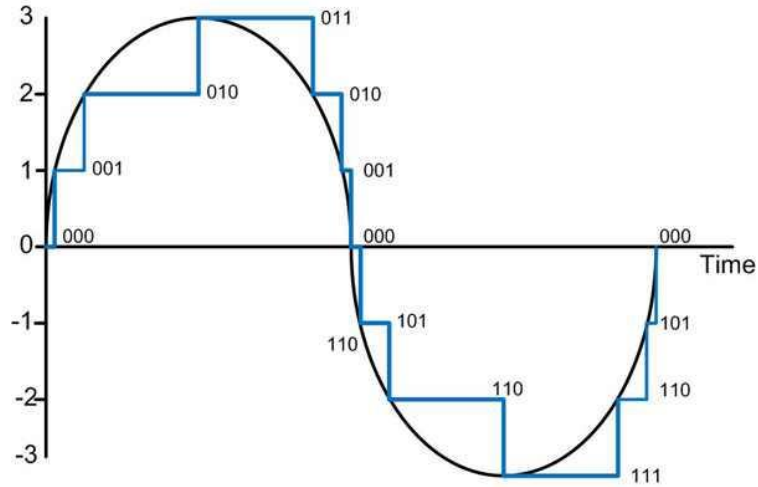
- ▶ Design a VI that generates a random number (0-100). Use shift registers with For Loop (20 iterations) to get the maximum of the generated 20 random numbers.

Note:

- ▶ Add Indicators for both the random number and the maximum number. Also add a delay (Wait function).
- ▶ You will have to use select function (Functions Palette >> Comparison>>Select)



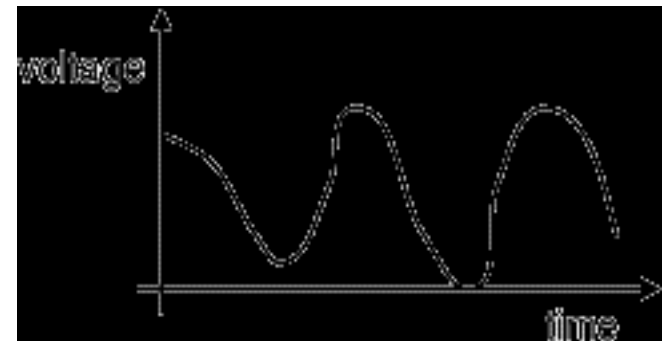
# Signals



# What is signal?

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- ▶ A *signal* is any kind of physical quantity that conveys information.



# Analog Signal:

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An ***analog signal*** is a kind of signal that is continuously variable.

- ▶ Temperature
- ▶ Pressure
- ▶ Position
- ▶ Speed
- ▶ Acceleration
- ▶ Sound
- ▶ Light Level

plus many others

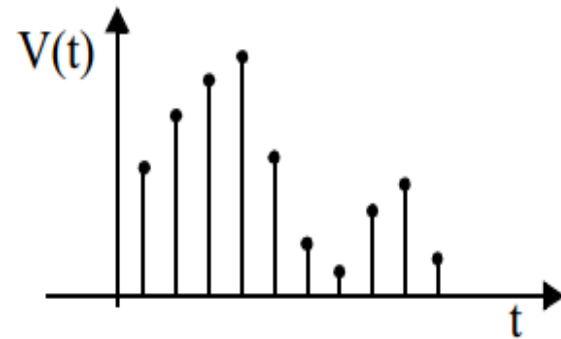
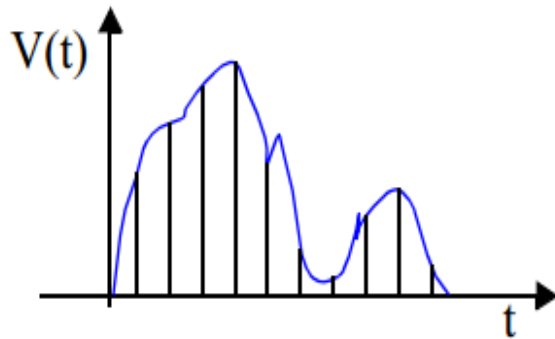
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# Digital Signal:

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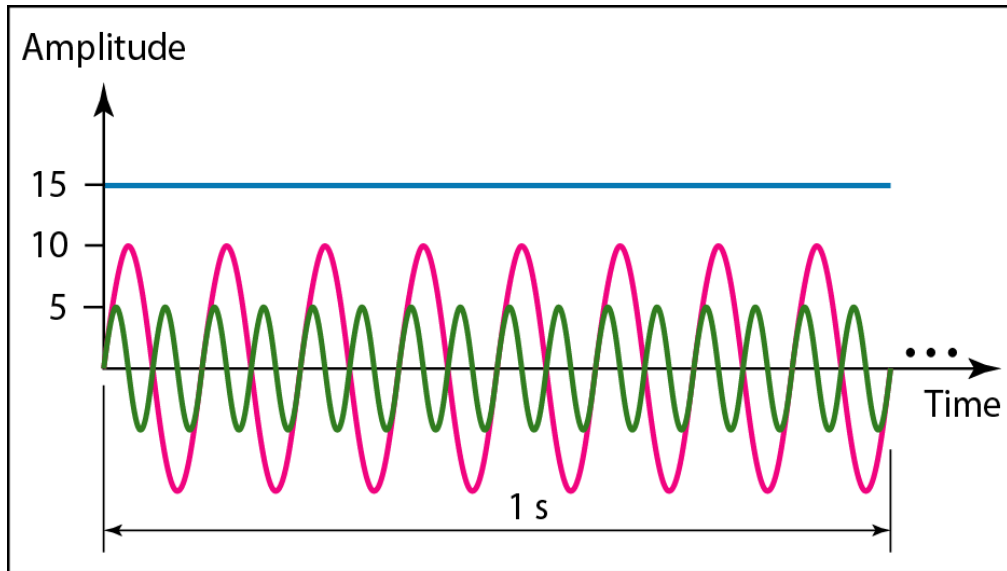
- ▶ A digital signal can have only a limited number of values.



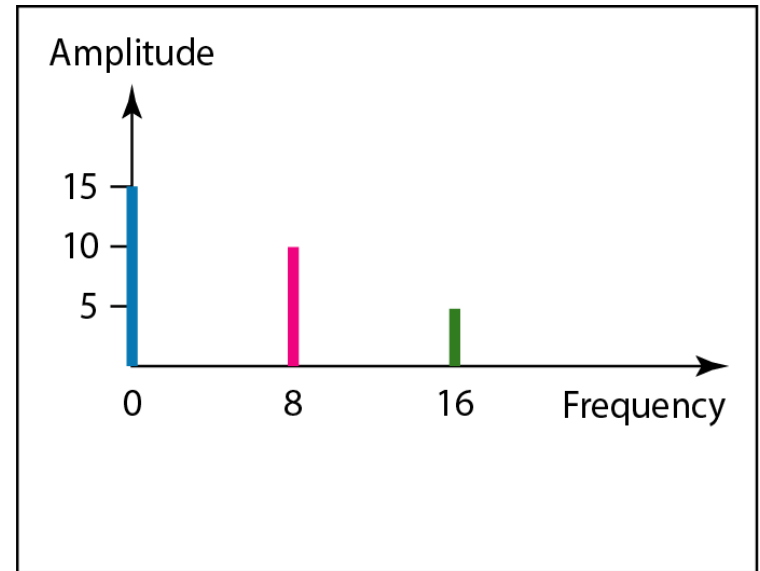
# Frequency Domain Analysis:

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- ▶ The frequency domain is more compact and useful when we are dealing with periodic functions.



a. Time-domain representation of three sine waves with frequencies 0, 8, and 16

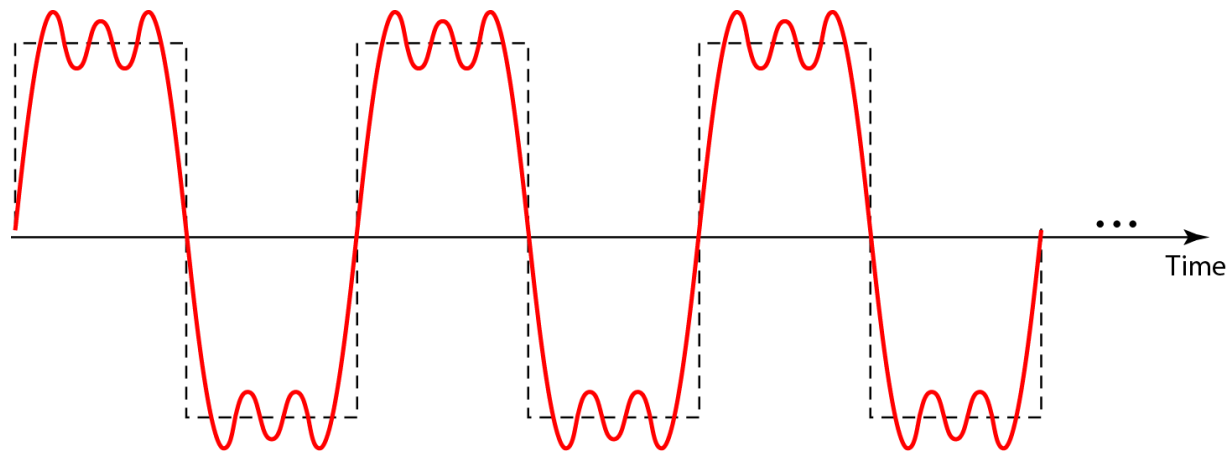


b. Frequency-domain representation of the same three signals

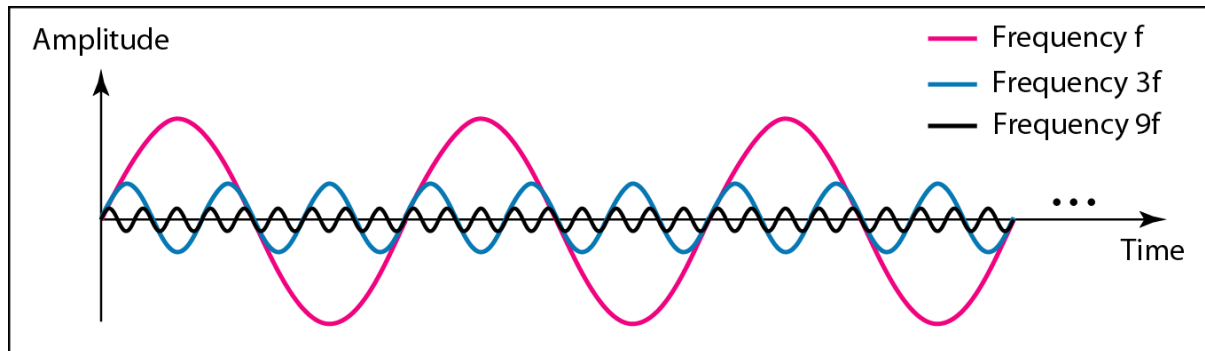
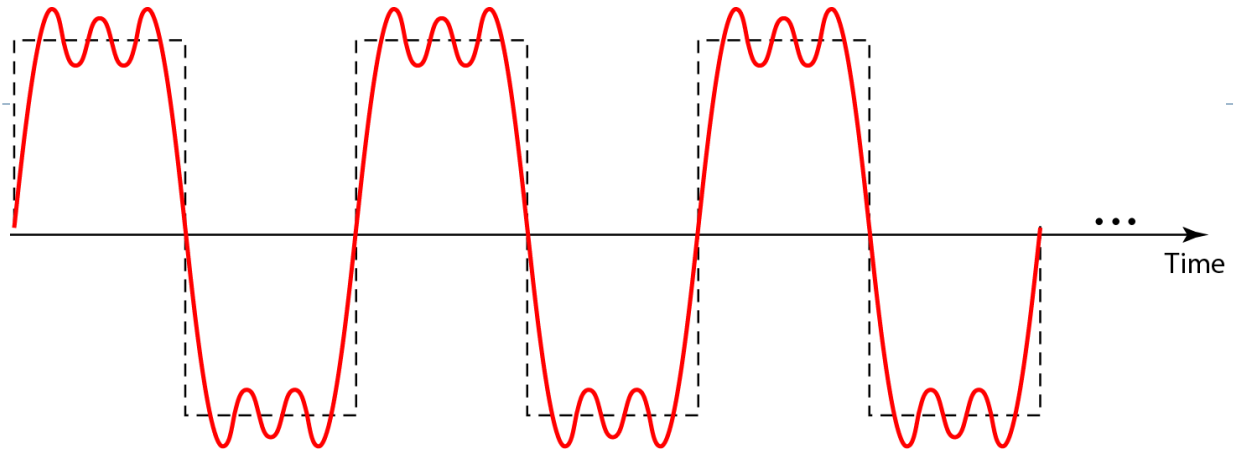




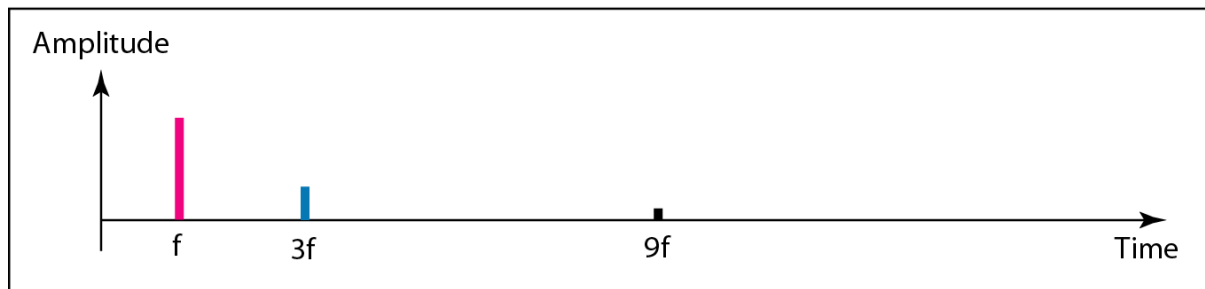
- 
- ▶ According to Fourier analysis, any composite signal is a combination of simple sine waves with different frequencies, amplitudes, and phases.



# Example:



a. Time-domain decomposition of a composite signal



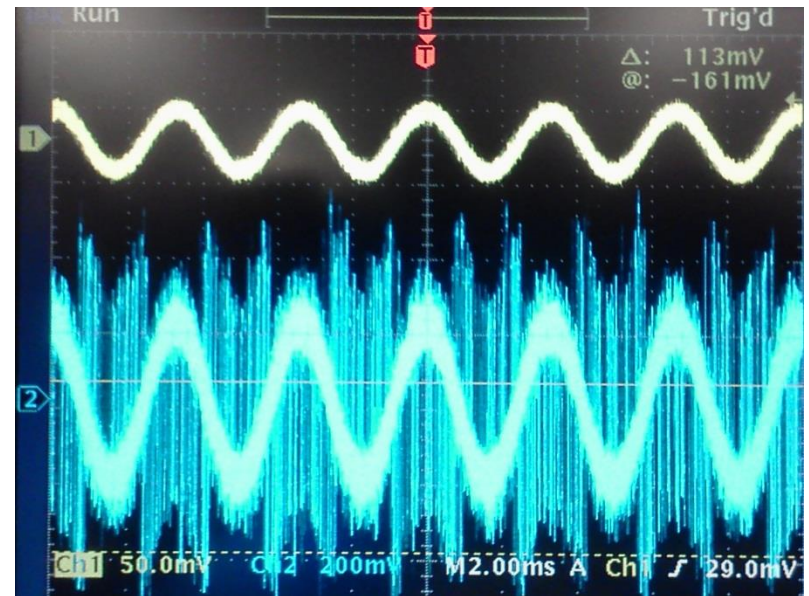
b. Frequency-domain decomposition of the composite signal

# Filters

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## What is a Filter?

- ▶ A *filter* is a system/process that processes a signal in some desired fashion.
- ▶ Filtering is a frequency selective operation, in which some frequency components are suppressed and some others are passed.



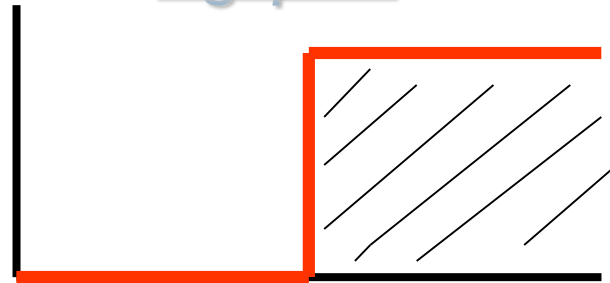
Background:

Four types of filters - "Ideal"

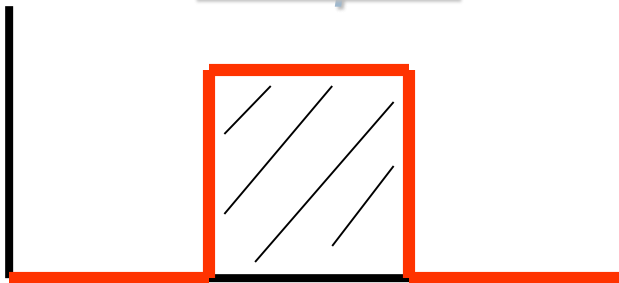
lowpass



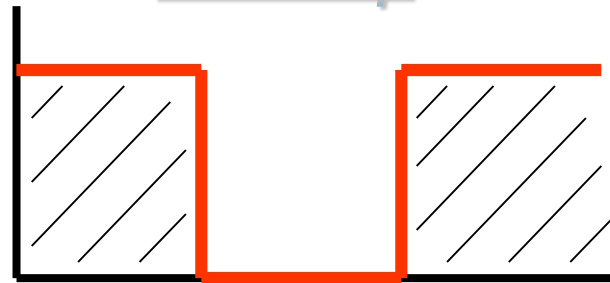
highpass



bandpass



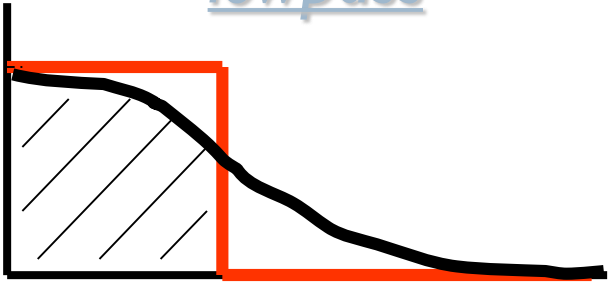
bandstop



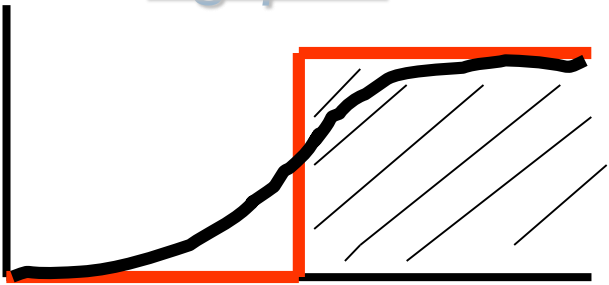
Background:

Realistic Filters: 

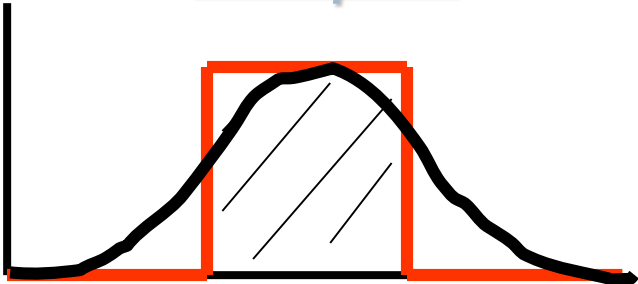
lowpass



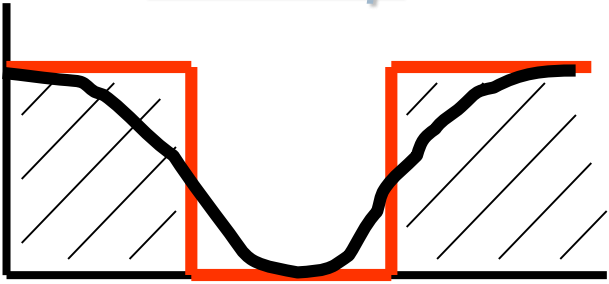
highpass



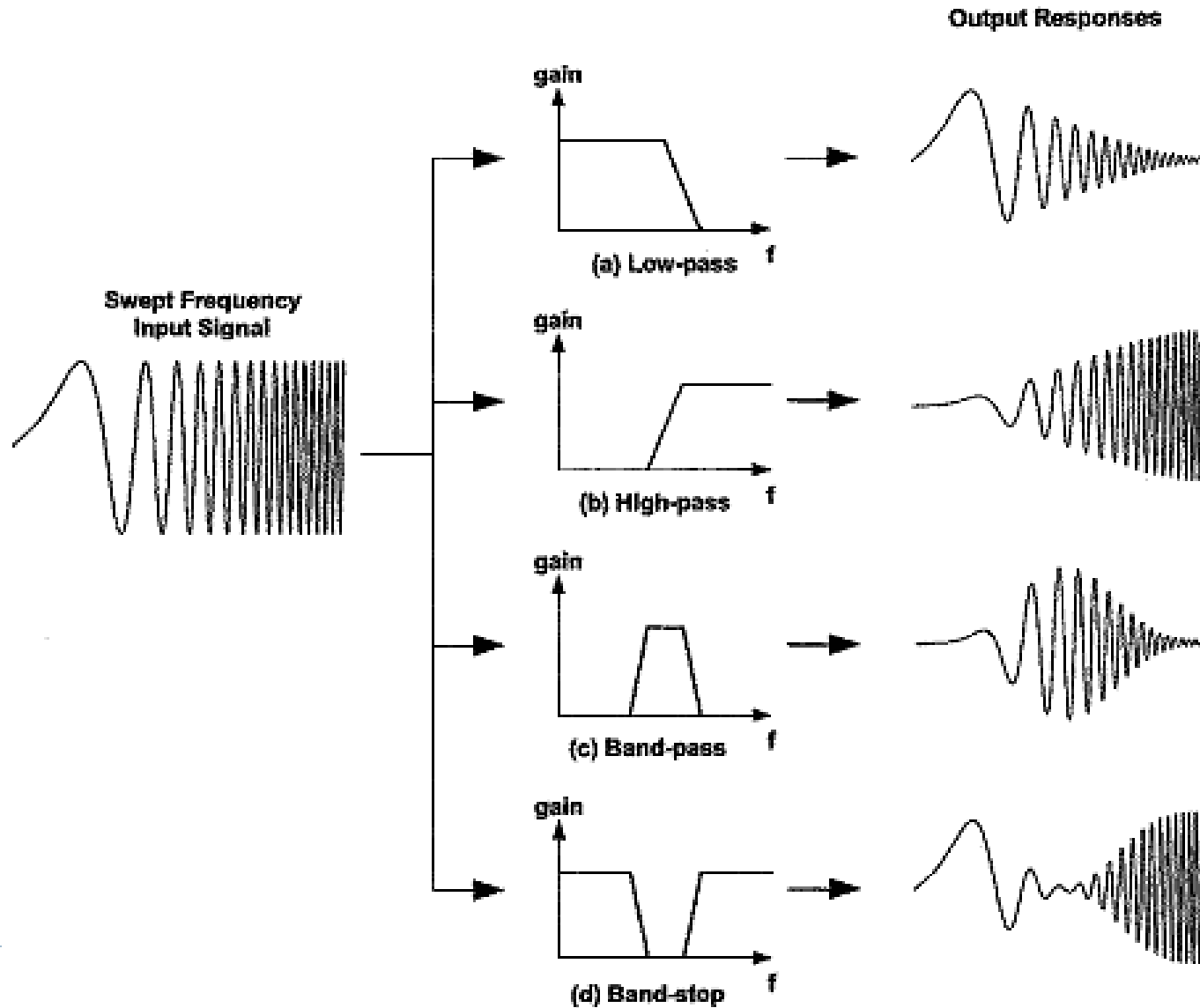
bandpass



bandstop



# Example



# Filters in LabVIEW

The image displays the LabVIEW Filter VI interface, which is used for filtering time signals. On the left, a small icon of the Filter VI is shown with its input and output ports. The main window, titled "Filter", contains the following elements:

- Input Signals:** A plot showing a noisy red signal with an amplitude range from -2 to 2 EU over a time range from 0 to 0.1.
- Output Signals:** A plot showing a smooth blue signal with an amplitude range from -500m to 1 EU over the same time range.
- Configuration Panel:** Located at the bottom, it includes:
  - Filter Specifications:** Mode (IIR Filter), Type (Lowpass), Topology (Butterworth), Order (2), and Cutoff (Hz) (20,000).
  - Filter Magnitude Response (dB):** A plot showing the magnitude response of the filter, with magnitude in dB on the y-axis (ranging from -100 to 20) and frequency in Hz on the x-axis (logarithmic scale from 2 to 500).
- Help Panel:** On the right, it provides a description of the Filter VI, its details, and feedback options.

## Exercise (3)

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Design a VI that generates three simulated sine-wave signals with user-controlled amplitude and frequency Knob. The VI adds the three sine wave signals and perform a filter operation. Show both signals (Original and the filtered one) on a single waveform chart

### **Notes:**

- ▶ Vary the signals frequencies and also the filter type (Low Pass - High Pass- Band Pass) to see the effect of these variations on the filtered signal.
  - ▶ Vary the filter characteristics such as order and cut off frequency, and observe the filtered signal.
- 





# Exercise (3)

