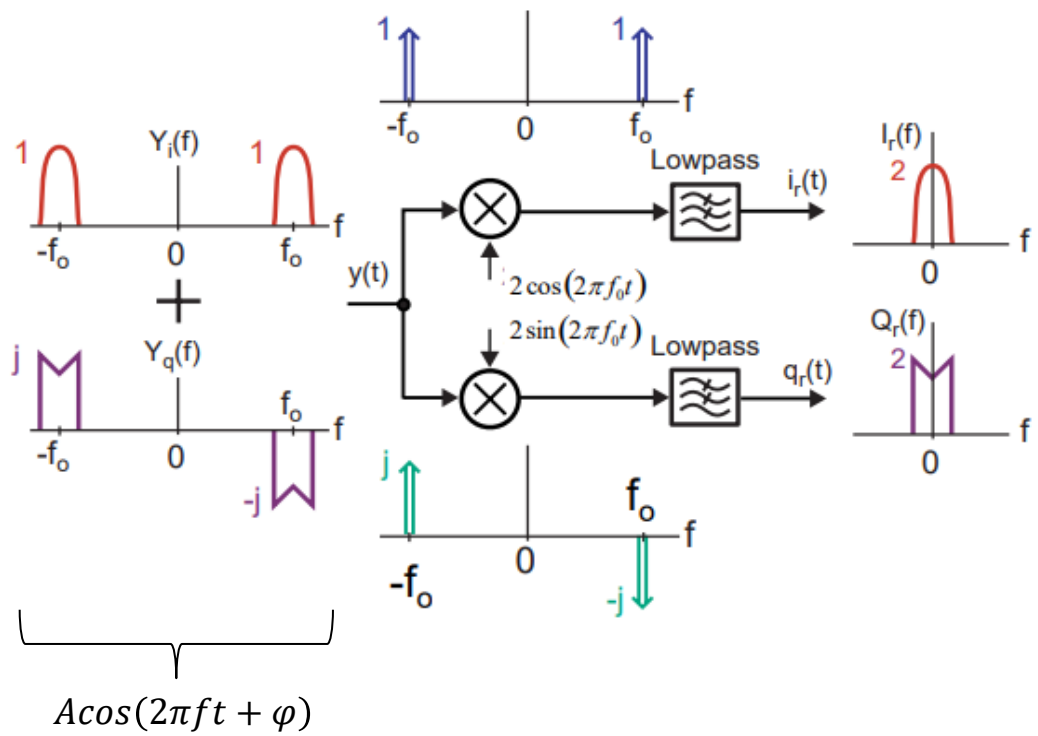


# IQ Demodulation



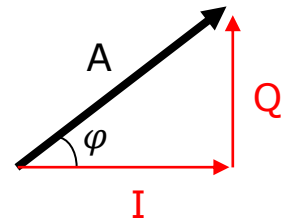
Important Trigonometric Identities:

$$\begin{aligned} \cos(a+b) &= \cos(a)\cos(b) - \sin(a)\sin(b) \\ \cos(a)\cos(a) &= 1/2(\cos(0) + \cos(2a)) \\ \sin(a)\cos(a) &= 1/2(\sin(2a) + \sin(0)) \\ \sin(a)\sin(a) &= 1/2(\cos(0) - \cos(2a)) \end{aligned}$$

$A \cos(2\pi ft + \varphi)$  is our signal that we want to reconstruct

$$A \cos(2\pi ft + \varphi) = A \cos(2\pi ft) \cos(\varphi) - A \sin(2\pi ft) \sin(\varphi)$$

As known:  
 $A \cos(\varphi) = I$   
 $A \sin(\varphi) = Q$



$A \cos(2\pi ft + \varphi) = I \cos(2\pi ft) - Q \sin(2\pi ft)$  is our signal

Lets multiply our signal with **LO**:  $2 \times \cos(2\pi ft)$  and  $2 \times \sin(2\pi ft)$

$$\begin{aligned} &2 \times A \cos(2\pi ft + \varphi) \times \cos(2\pi ft) \\ &= 2 \times (I \cos(2\pi ft) - Q \sin(2\pi ft)) \times \cos(2\pi ft) \\ &= I \times (\cos(0) + \cos(4\pi ft)) - Q \times (\sin(4\pi ft) + \sin(0)) \\ &= I + I \times \cos(4\pi ft) - Q \times \sin(4\pi ft) \end{aligned}$$

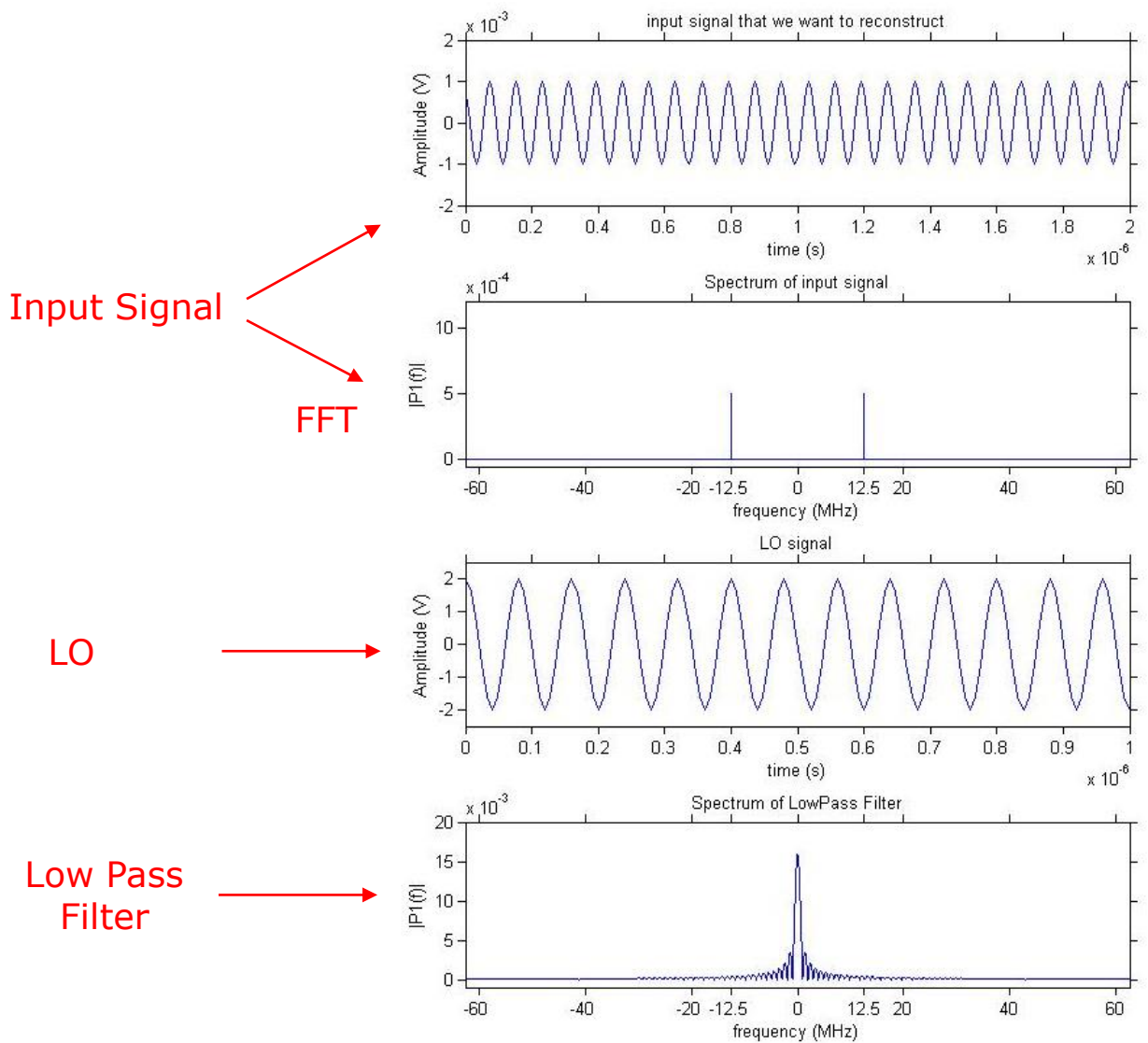
$$\begin{aligned} &2 \times A \cos(2\pi ft + \varphi) \times \sin(2\pi ft) \\ &= 2 \times (I \cos(2\pi ft) - Q \sin(2\pi ft)) \times \sin(2\pi ft) \\ &= I \times (\sin(4\pi ft) + \sin(0)) - Q \times (\cos(0) - \cos(4\pi ft)) \\ &= I \times \sin(4\pi ft) - Q + Q \times \cos(4\pi ft) \end{aligned}$$

*If you have I and Q, you can calculate A and φ, and reconstruct your signal*

$$A(t) = \sqrt{I(t)^2 + Q(t)^2} \qquad \varphi(t) = \tan^{-1} \frac{Q(t)}{I(t)}$$

$$A(t) \cos(2\pi ft + \varphi(t))$$

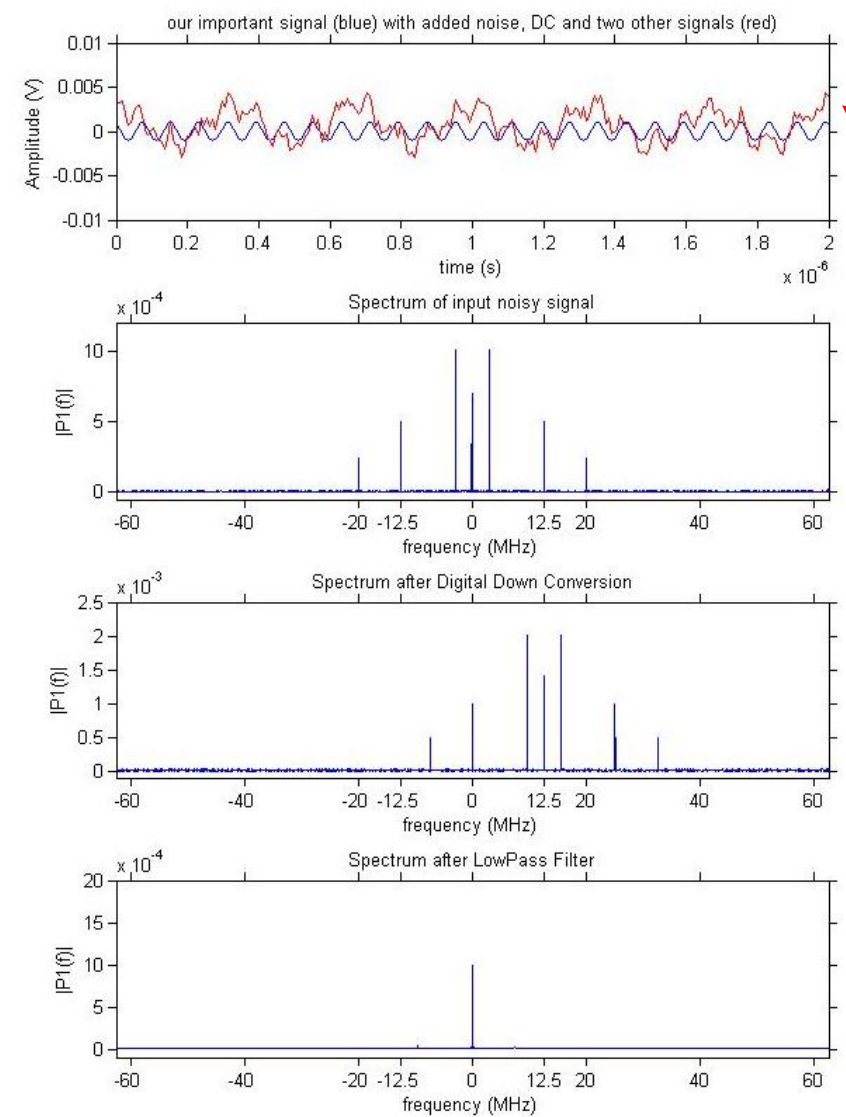
# Project Goal (Lockin\_amplifier\_solution.m)



Input Signal  
FFT

LO

Low Pass Filter



Input Signal with Added Signals and noise  
FFT

After DDC

FFT of I & Q

# Project Goal

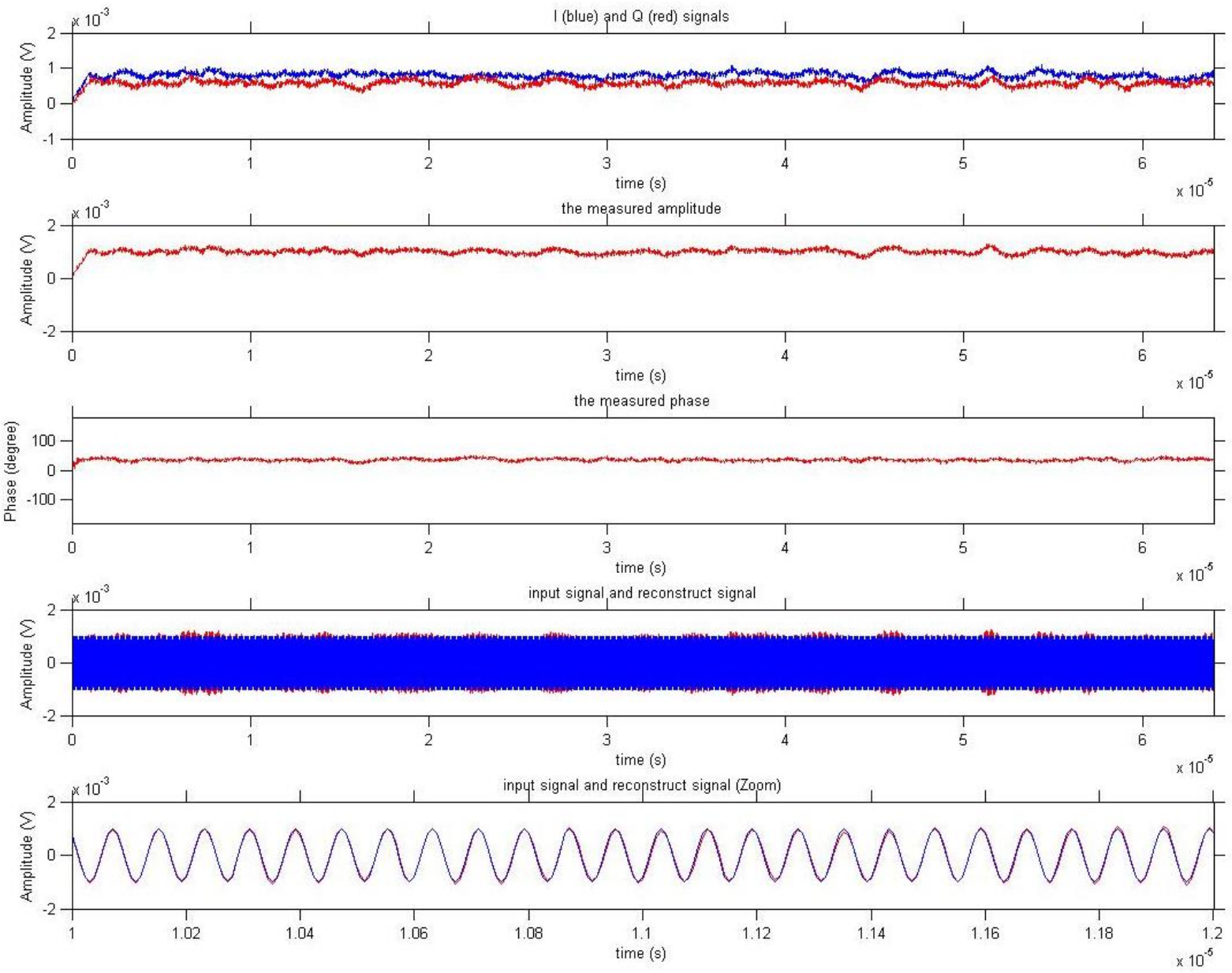
I & Q

Amplitude

Phase

Reconstructed Signal

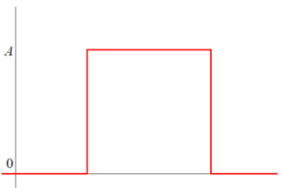
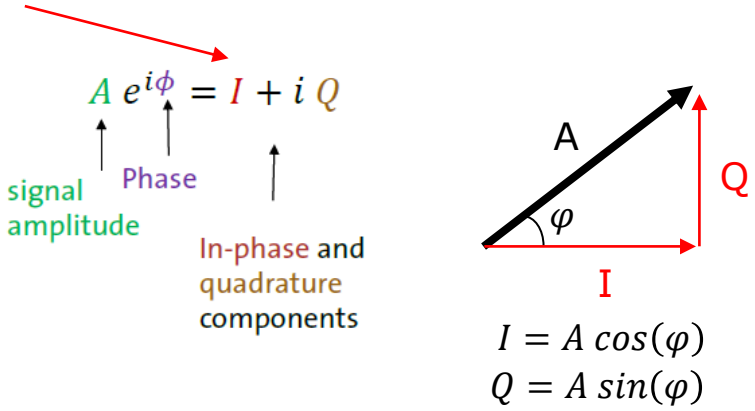
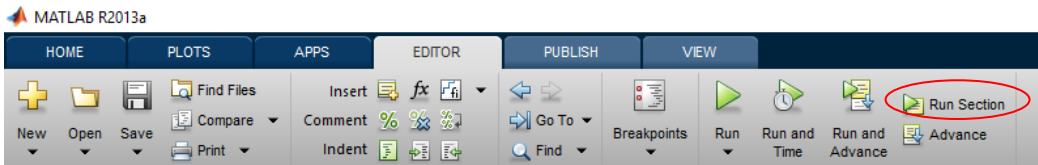
Reconstructed Signal (Zoomed)



# Lab Session: IQ Demodulation to measure the phase and amplitude of a signal

## Step by Step:

- Open `lockin_amplifier.m` file with matlab
- Search `xxxxx` in the file to see what you need to add or change.
- Check how our input signal `signal_in` is generated (lines 7-17). What is its phase (in degree and radian) ?
- Take the `fft` of `signal_in` (line 29). **Run Section.**
- Generate `signal_s1` and `signal_s2`, which are added on top of `signal_in`. Use `t_range` to define time interval:
  - signal\_s1: 3 MHz cos signal with 0.002 V amplitude, 0 radian phase
  - signal\_s2: 20 MHz sin signal with 0.0005 V amplitude, pi/4 radian phase
- Take the `fft` of `signal_in_noisy` (line 64)
- Apply `fftshift` (line 65). Normalize `fftshift` value with number of samples (check line 30 as an example). `fftshift` puts 0 Hz to the middle, otherwise 0 Hz is in the left edge, which is not conventional for `fft` visualization. **Run Section.**
- Define your LO signals according to LO signal definitions at slide 16. (line 79-83)
- Multiply `signal_in_noisy` with LO signals (lines 85-86). Now you implemented digital down conversion (DDC).
- Use `I_before_filter` and `Q_before_filter` to define the signal with inphase and quadrature. Formula is here.
- **Run Section.** Check if the frequency components of `signal_in_noisy` are shifted after DDC.
- Check lines 113-120. It is a boxcar filter (averager of consecutive values) and the output is I and Q values. It filters out high frequency components. Check Slide 16. After boxcar filter for Q value will you get Q or -Q? According to that, multiply your filtered signals with 1 or -1 (Line 116-117).
- **Run Section.** Check the plots titled as "Spectrum after LowPass Filter" and "Spectrum of Low Pass Filter". Do you think is boxcar perfect as a low pass filter, why or why not ?



Boxcar filter in time domain

# Lab Session: IQ Demodulation to measure the phase and amplitude of a signal

## Step by Step:

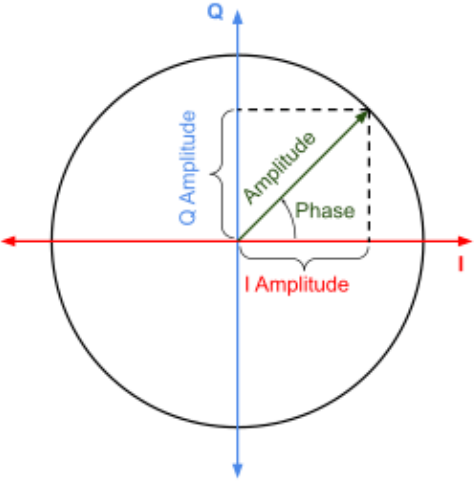
- Now we have I and Q components. We can extract amplitude and phase from I and Q. Then we can reconstruct our signal. *A\_reconstruct* is the amplitude, *P\_reconstruct* is the phase. Calculate them using the formulas below. Use *atan2(y,x)* function of matlab to calculate the arctangent. (Line 150-151)

$$A(t) = \sqrt{I(t)^2 + Q(t)^2} \qquad \varphi(t) = \tan^{-1} \frac{Q(t)}{I(t)}$$

- Reconstruct the signal using the formula below. (Line 153)

$$A(t) \cos(2\pi ft + \varphi(t))$$

- **Run Section.**
- Is the reconstruction successful ? Check the phase and amplitude plots. Is it matching with your input signal's amplitude and phase ?
- Change the phase of your input signal *signal\_in* between 0 and  $2\pi$ , and check if the reconstruction is correct.
- Check the plots of I and Q values and the Phase. Considering that I and Q values are positive or negative, phase can be at one of the four regions around the sphere (0 to  $2\pi$ ). Does it match in your plot (for example when both I and Q are negative, phase should be between  $\pi$  and  $1.5\pi$ )?



# Lab Session: IQ Demodulation to measure the phase and amplitude of a signal

## Step by Step:

- By looking at I and Q components and their region at complex plane, we can distinguish two qubit states ( $e$  and  $g$ ) after measurement.
- measurements.mat file includes 3 measurement signals. Type clear all in console. Click [measurements.mat](#). You will see 3 data sequence in workspace. You expect to see your quantum signal at 12.5 MHz after analog down conversions.
- In line 17, type `signal_in=0;`
- Comment out `clear all` at line3 (to dont click [measurements.mat](#) file every time)
- In line 17, type: `signal_in=0;`
- In line 48, type: `signal_in_noisy = measurement_1;`
- Run all code, and repeat the runs by assigning `measurement_2` and `measurement_3` to `signal_in_noisy`.
- Answer the questions for each run, considering that you expect a response from your qubits as given in figures below:

When do the measurements start and stop ?  
What is the qubit's state for signal 1 & 2 ?  
What happens at  $2.8e-5$  s in signal 3 ?

