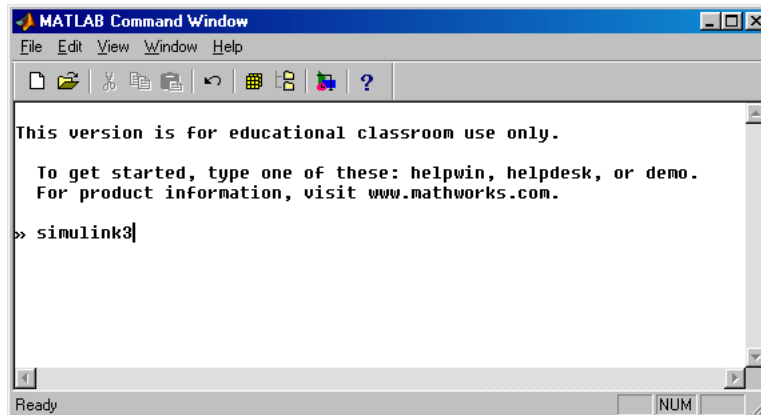
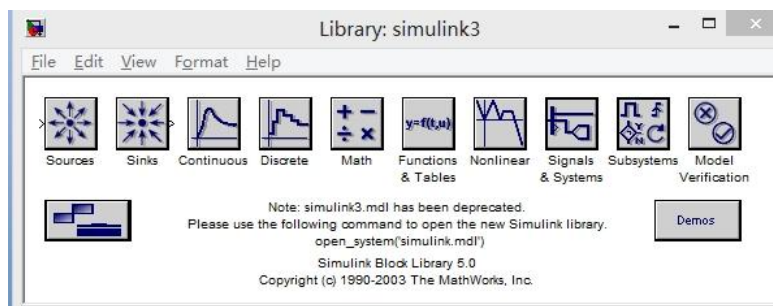


## Simulink – Discrete Fourier Transform

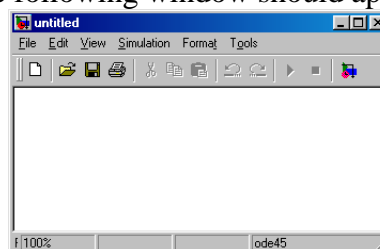
1. Start the *Matlab* engine and type *simulink3* at the Matlab command prompt. This will start the Simulink3 library.



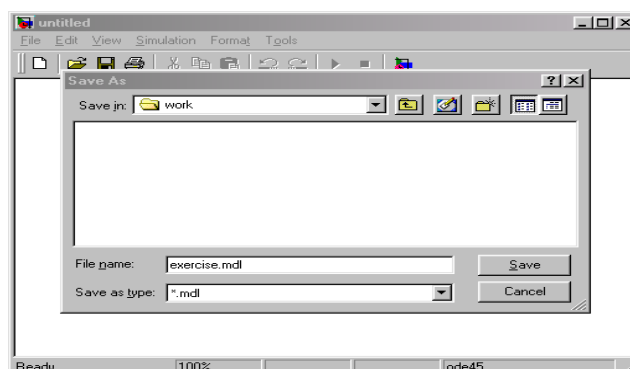
2. To open a new model select: File → New → Model



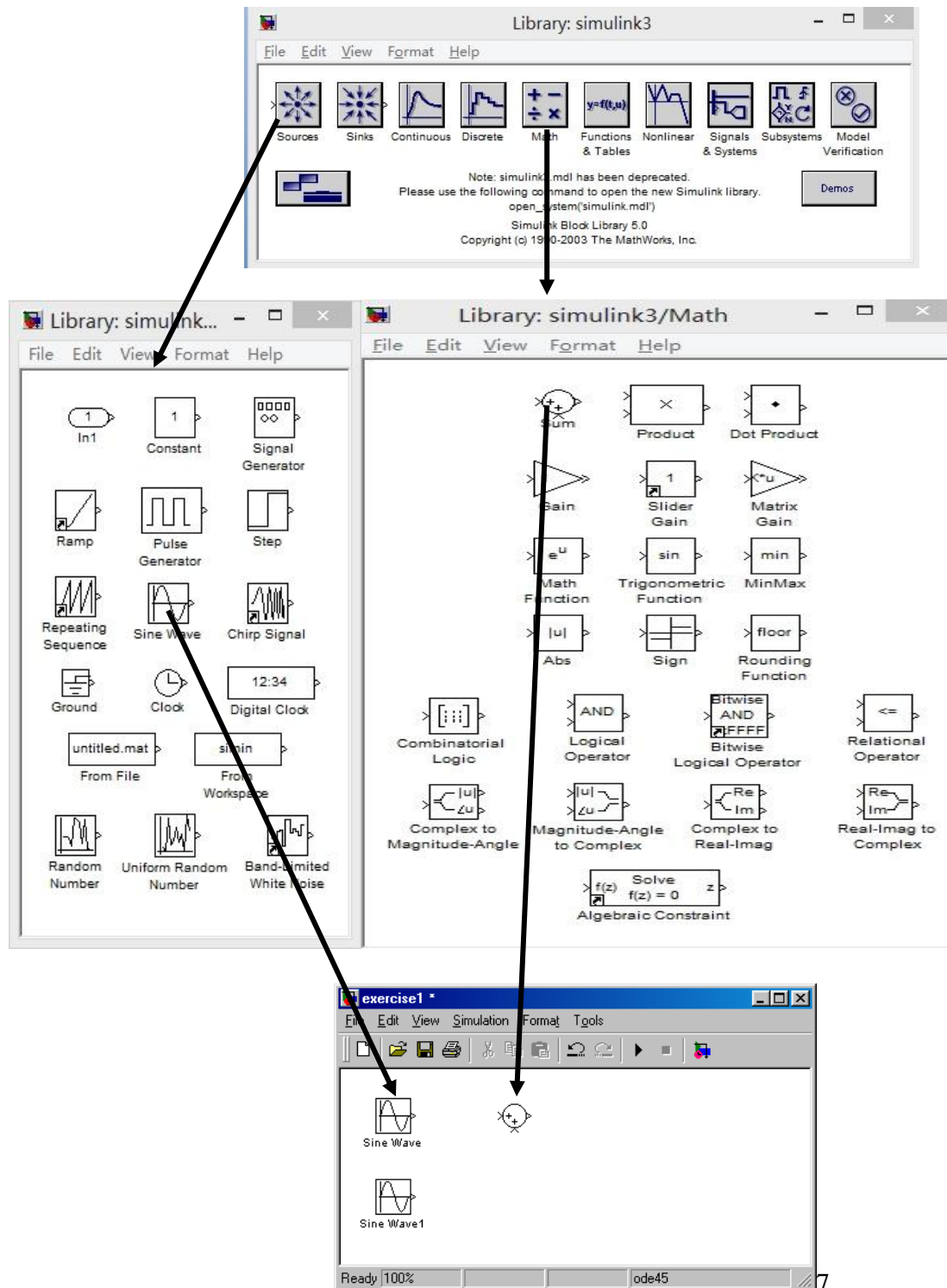
(The following window should appear.)



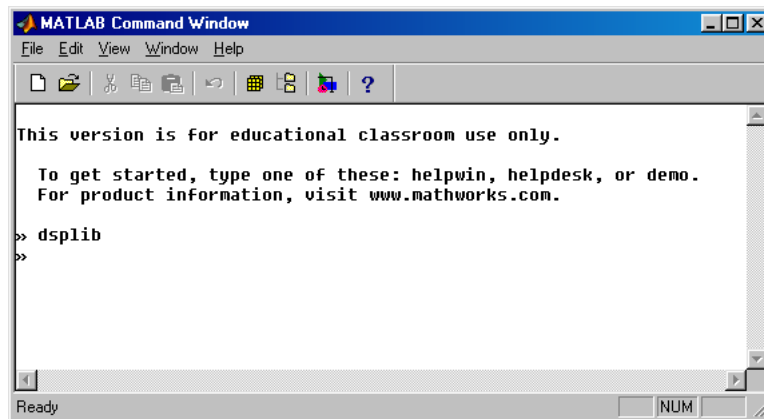
3. To save the new simulink model select: File → Save As



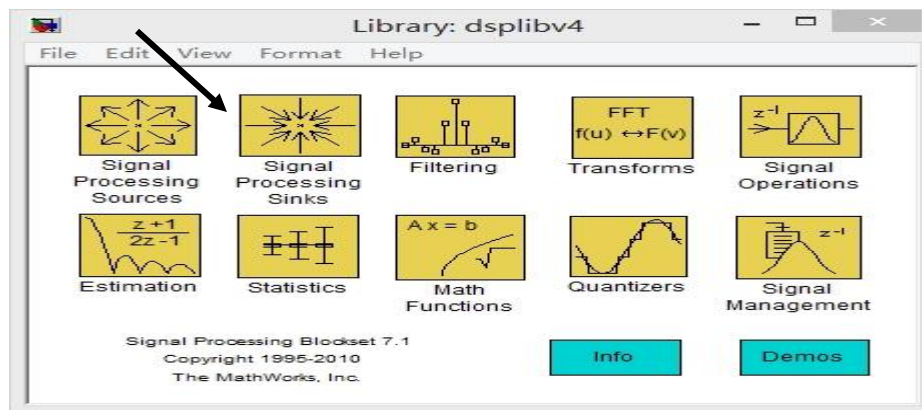
4. Double click Sources, Drag & drop two sine-wave blocks and a summation block from Math icon of the simulink3 libraries as illustrated below.



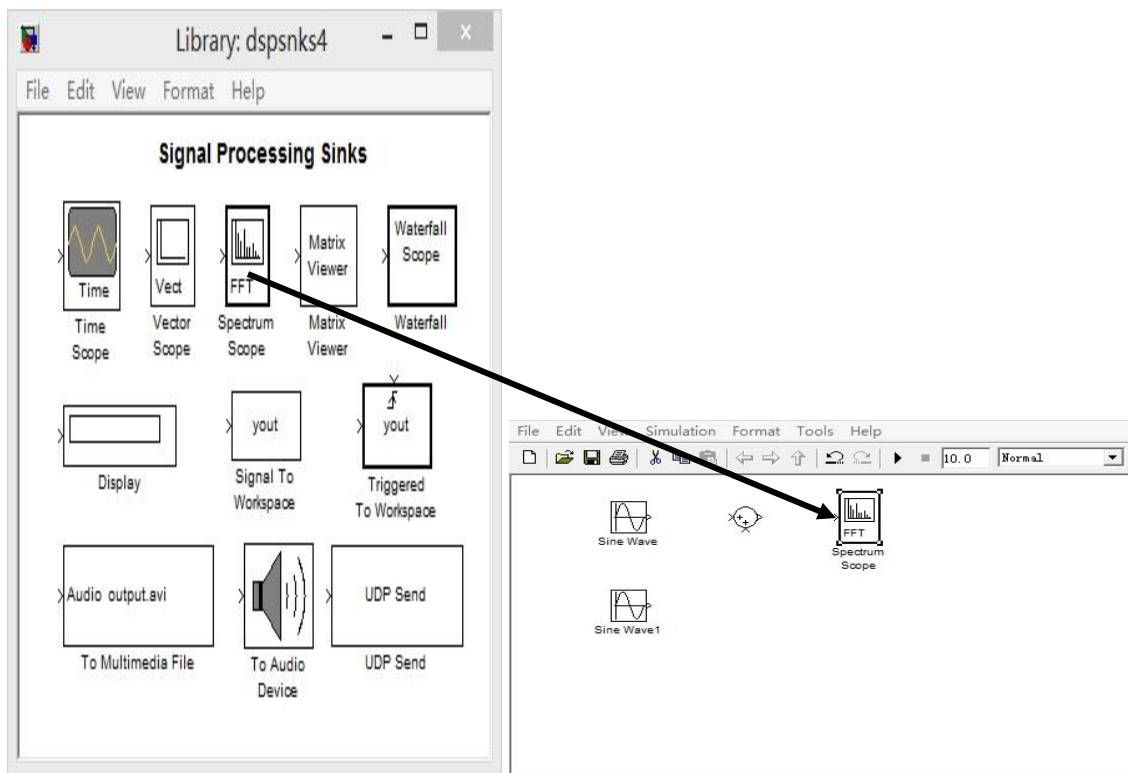
5. Open the DSP Library by typing *dsplib* at the command prompt.



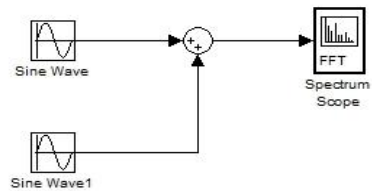
6. Open the DSP Sinks library by double-clicking the *Signal Processing Sinks* icon.



7. Drag & drop the *FFT Spectrum Scope* block.



8. Connect the simulink blocks.



9. To enter the sine-wave block parameters double-click the corresponding icon.

Recall:  $\text{Amplitude} * \sin(2 * \pi * \text{Frequency} * t + \text{Phase})$   
 where  $t = 0, T_s, 2 * T_s, 3 * T_s, \dots$

Source Block Parameters: Sine Wave

Use the sample-based sine type if numerical problems due to running for large times (e.g. overflow in absolute time) occur.

Parameters

Sine type: Time based

Time (t): Use simulation time

Amplitude: 1

Bias: 0

Frequency (rad/sec):  $2\pi \cdot fc1$

Phase (rad): 0

Sample time: Ts

☒ Interpret vector parameters as 1-D

OK Cancel Help

10. Repeat above for the block sine-wave1 (careful of  $fc2$ )

Source Block Parameters: Sine Wave1

Use the sample-based sine type if numerical problems due to running for large times (e.g. overflow in absolute time) occur.

Parameters

Sine type: Time based

Time (t): Use simulation time

Amplitude: 1

Bias: 0

Frequency (rad/sec):  $2\pi \cdot fc2$

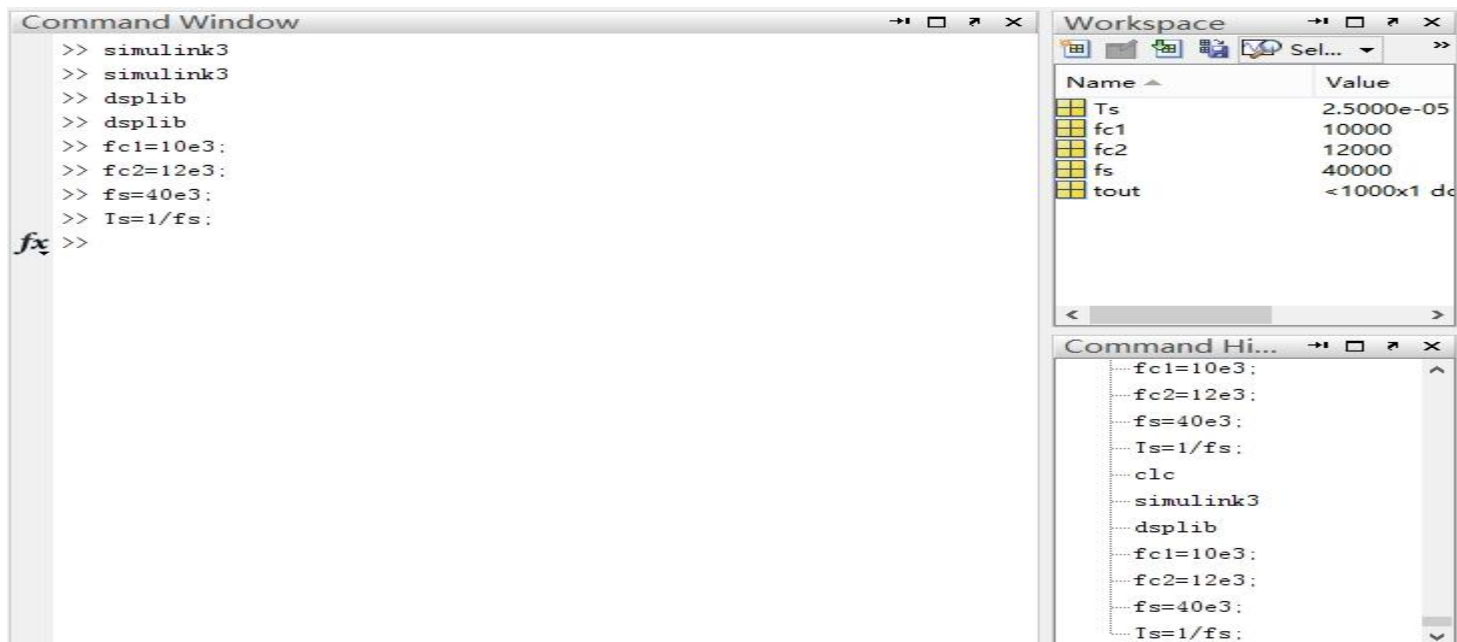
Phase (rad): 0

Sample time: Ts

☒ Interpret vector parameters as 1-D

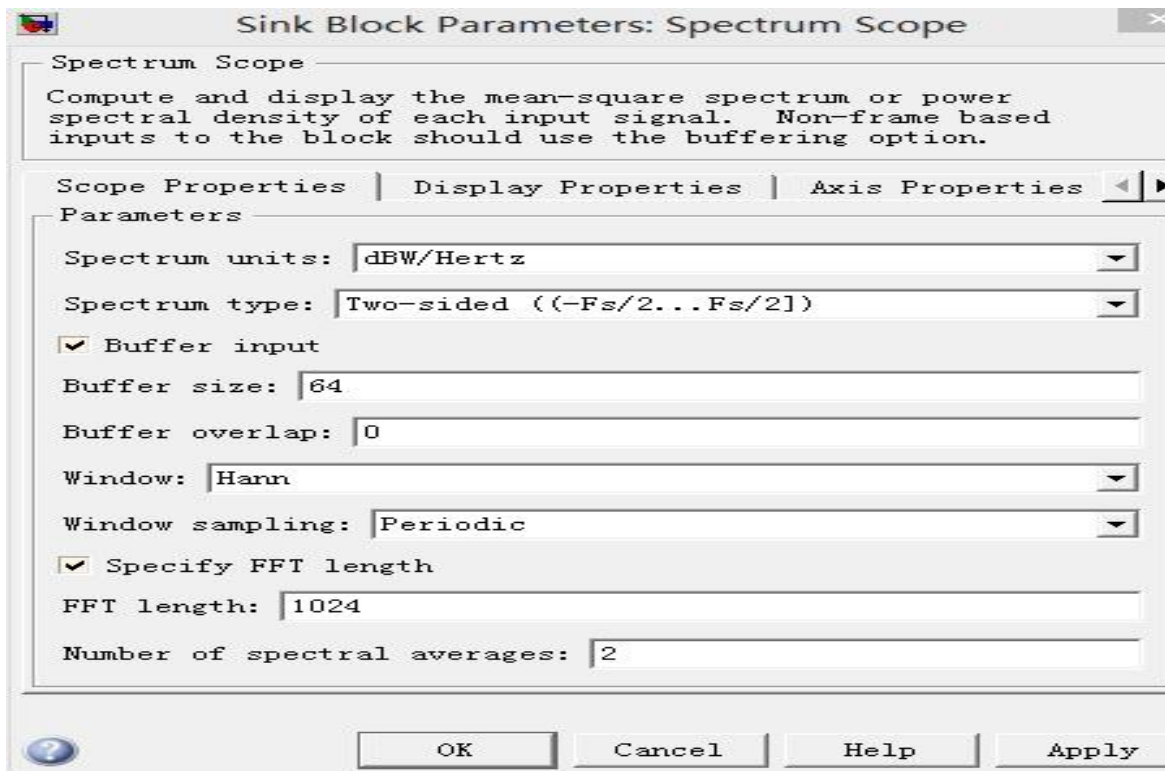
OK Cancel Help

11. Specify  $f_{c1}$ ,  $f_{c2}$ ,  $f_s$ , and  $T_s$  at the command prompt.

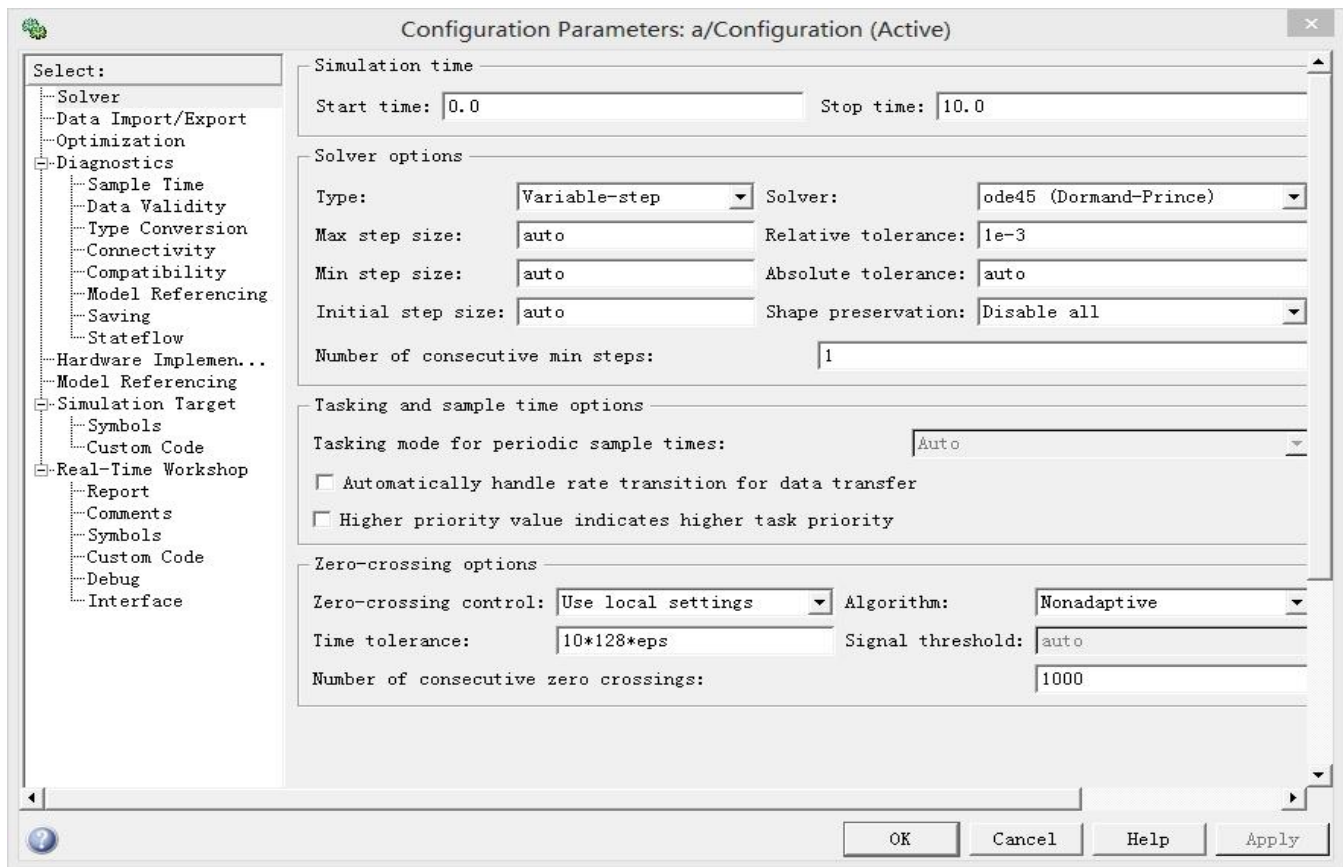


where  $f_{c1}$  and  $f_{c2}$  are the carrier frequencies,  $f_s$  is the sampling frequency, and  $T_s$  is the sampling time.

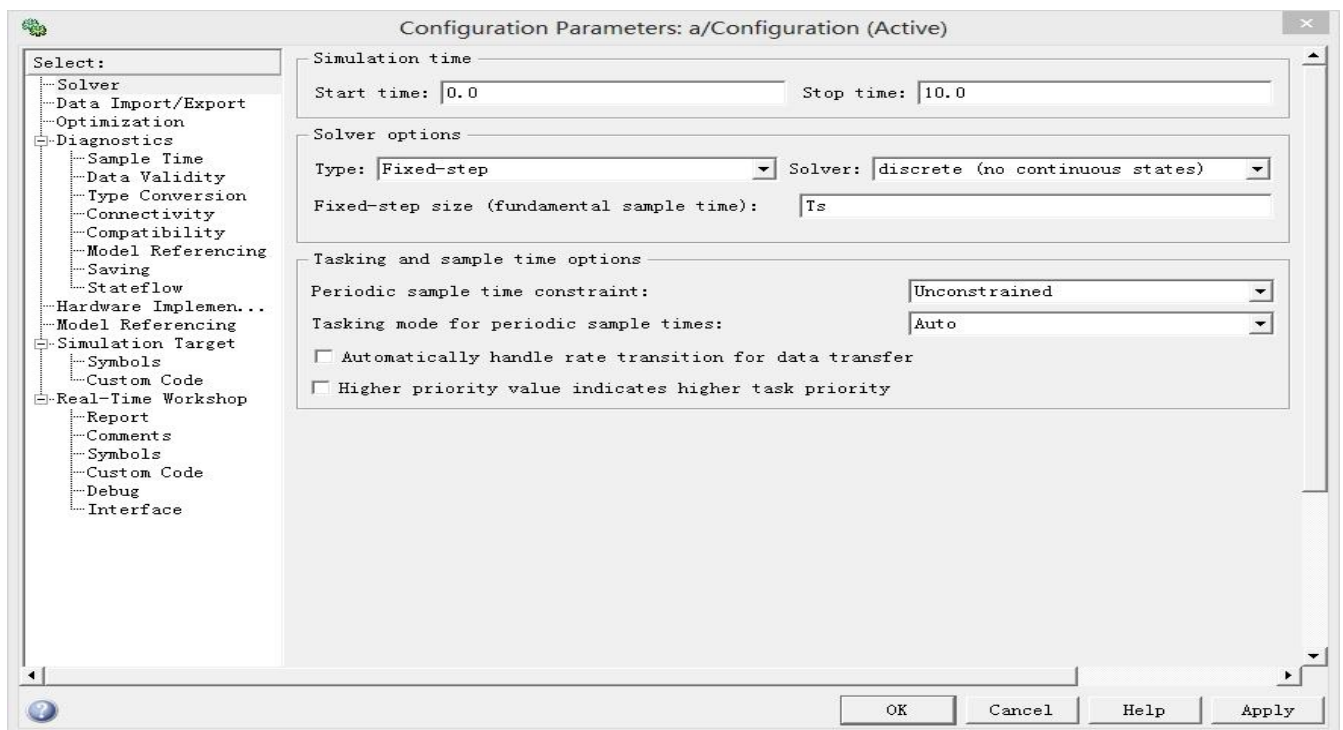
11. Enter the following *FFT scope* parameters.



12. Finally, to enter the Simulation Parameters choose: Simulation → Configurations Parameters.

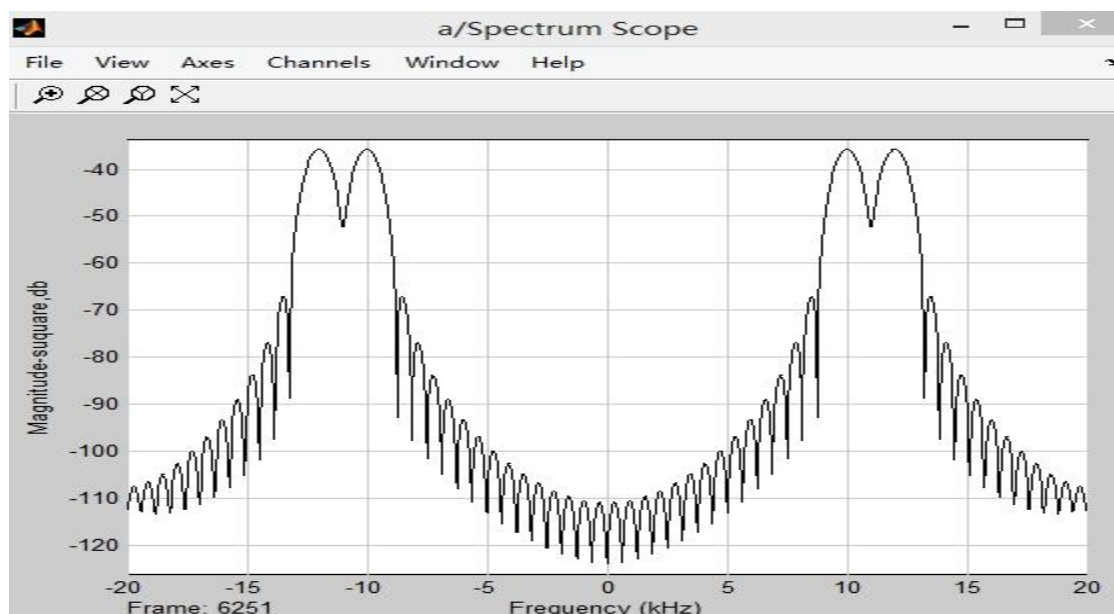


From Type: choose Fixed-step and follow the below setting up and press Apply and click OK





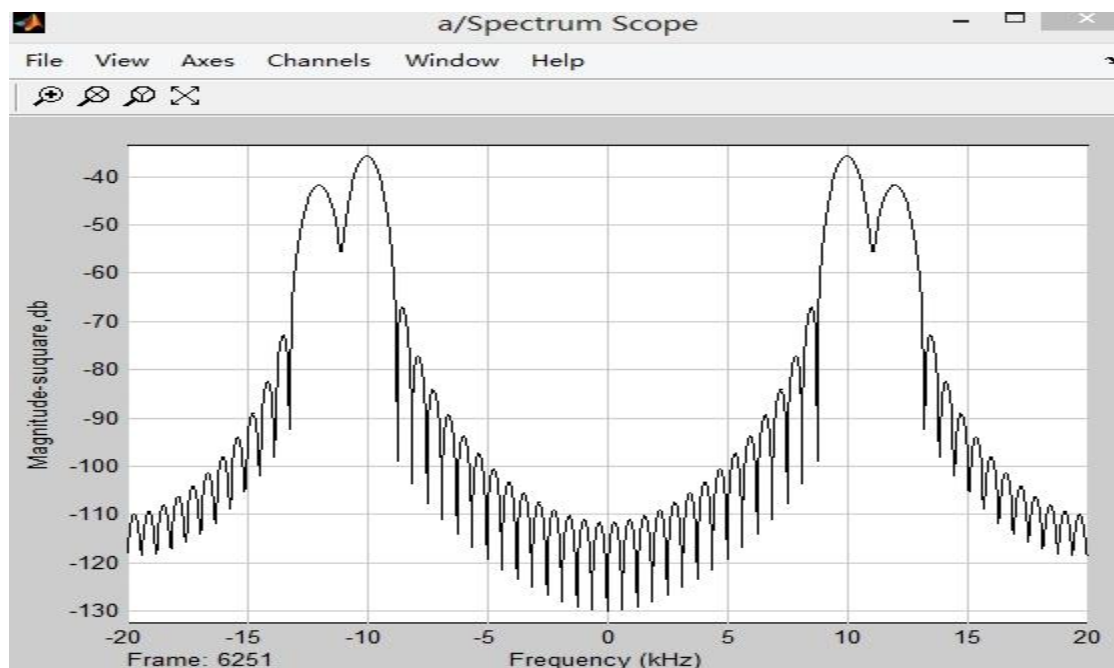
13. Choose simulation and click start or point & click at the play button on the toolbar.
14. Once the simulation is running right click on the scope window and choose autoscale. The following plot should appear.



Explain the resulting spectral components by considering the following Fourier transform pair:

$$\sin(2\pi f_c t) \xleftrightarrow{FT} \frac{1}{2j} \delta(f - f_c) - \frac{1}{2j} \delta(f + f_c)$$

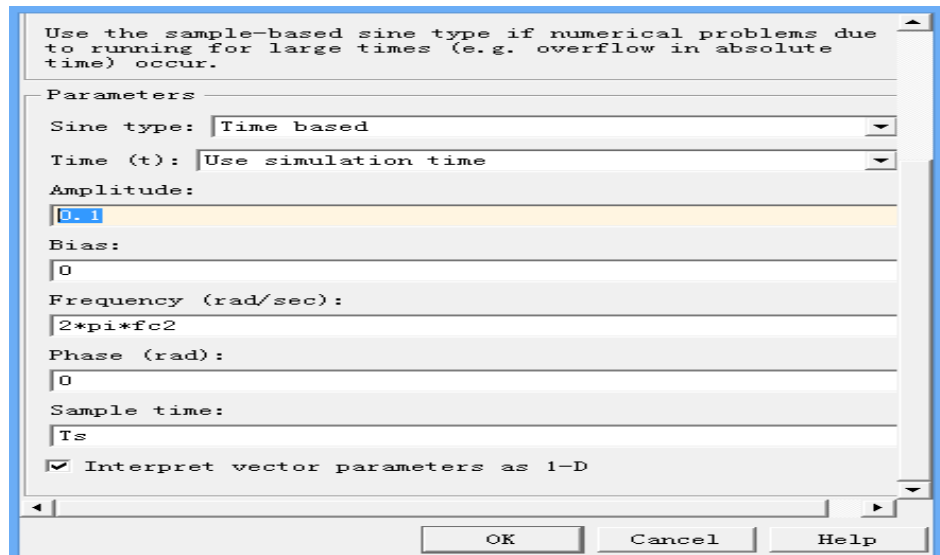
15. Change the amplitude of the sine-wave1 to 0.5, save the changes, and re-run the simulation.



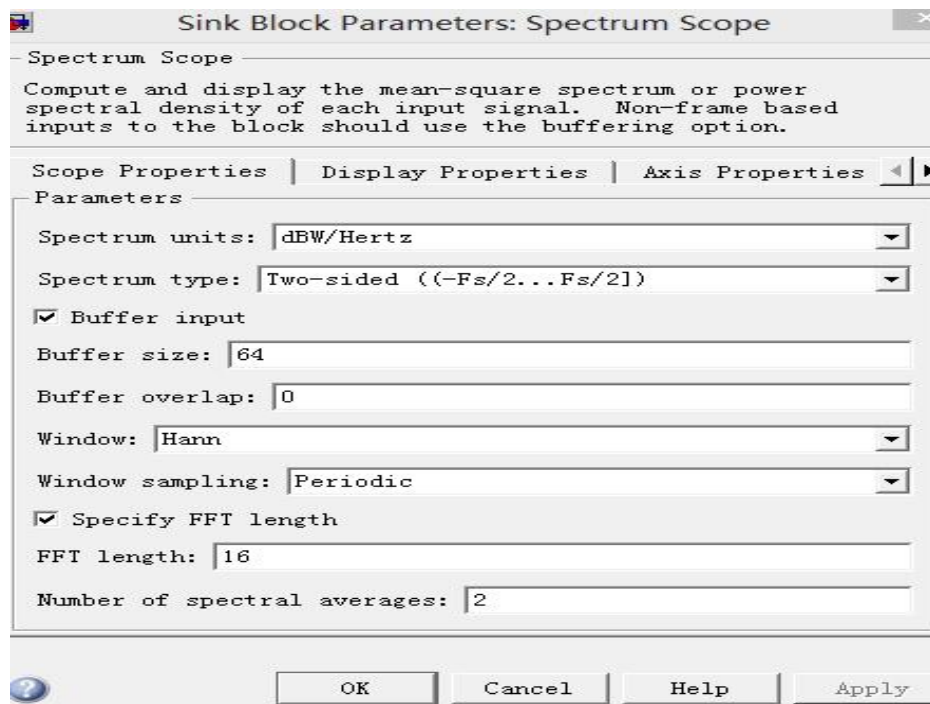


Explain the resulting spectrum.

- Change of the sine-wave1 to 0.1 and number of FFT to  $N=16$  which is expressed as follow, explain the results



and



- Change number of FFT back to  $N=1024$ , re-run the simulation, explain the results.
- Change  $fc2=10500$ , amplitude of sine-wave1 is set to 1 and choose different window function, re-run the simulation, explain the results.

Sink Block Parameters: Spectrum Scope

Spectrum Scope

Compute and display the mean-square spectrum or power spectral density of each input signal. Non-frame based inputs to the block should use the buffering option.

Scope Properties | Display Properties | Axis Properties

Parameters

Spectrum units: dBW/Hertz

Spectrum type: Two-sided ( $(-F_s/2 \dots F_s/2)$ )

☒ Buffer input

Buffer size: 64

Buffer overlap: 0

Window: Chebyshev

Stopband Blackman  
Boxcar

☒ Specif Chebyshev

Hamming

FFT leng Harn

Hanning

Number o Kaiser

Triang

?

OK Cancel Help Apply