

## **Module Objectives:**

- 1. Review the decibel.
- 2. Discuss time-domain windowing and the window shape and length effects on FFT resolution.

#### **Reading for this Module**

- 1. DSP4 chapters 5.3 5.4, 5.6
- 2. Please read handout DSP\_Handout\_Digital\_Sinc.pdf
  There is also posted code with this handout: DSP\_Digital\_Sinc\_Example.m
- 3. Please review DSP\_Handout\_Common\_Filters\_Summary.pdf
- 4. Orfanidis handouts these go over the details behind the frequency resolutions DSP\_Handout\_Orfanidis\_Reading\_Material\_Part\_II.pdf DSP\_Handout\_Orfanidis\_Reading\_Material\_Part\_II.pdf

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# Module 3.3 - Frequency Analysis Part 3

## In-class computation:

Matlab and Excel.

**Canvas:** 

DSP\_FFT\_Resolution.m DSP\_FFT\_Resolution.fig

DSP\_FFT\_Resolution\_Group\_X.xlsx (Edit the one for your group).

You can run code ahead of time but let's not populate the XLSX sheet until we do exercises in class please.

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#### **Announcements**

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# **Module 3.3 - Frequency Analysis Part 3**

## The Decibel (dB)

dB

100

90

80

70

60

50

40

30

20

10

6

3

1

**Power Ratio** 

10,000,000,000:1

1,000,000,000:1

100,000,000:1

10,000,000:1

1,000,000:1

100,000:1

10,000:1

1,000:1

3.981:1

1.259:1

1.995 (~2):1

100:1

10:1

dB

-1

-3

-6

-10

-20

-30

- 40

-50

-60

**-70** 

-80

-90

-100

**Power Ratio** 

0.501 (~1/2):1

0.794:1

0.251:1

0.1:1

0.01:1

0.001:1

0.0001:1

0.000 01:1

0.000 001:1

0.000 000 1:1

0.000 000 01:1

0.000 000 001:1

0.000 000 000 1:1

The **Decibel** (dB) is used extensively in many signal processing applications due to the large dynamic ranges of signals involved. The dB is a non-linear, logarithmic scaling of power ratios:

$$X_{dB} = 10 \times log_{10}(x)$$

When dea magnitud ted

The term x is called a natural unit.

aling with <u>voltages or plotting filter</u> l <u>es</u> , we need power, so the dB is calculat		
$X_{dB} = 20 \times log_{10}(x)$		

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## The Decibel (dB)

When working with decibels: multiplications in natural units add in dB

divisions in natural units subtract in dB.

**Example:** Convert 400W to dB (without using a calculator).

$$10 \times log_{10}(400)$$

$$= 10 \times log_{10}(2 \times 2 \times 10 \times 10)$$

$$= 10 \times log_{10}(2) + 10 \times log_{10}(2) + 10 \times log_{10}(10) + 10 \times log_{10}(10)$$

$$= 3 + 3 + 10 + 10$$

$$= 26 \ dB$$

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# **Module 3.3 - Frequency Analysis Part 3**

## The Decibel (dB)

dB's are also easy to work with when the quantity is raised to a power:

Let

$$10 \times log_{10}(x) = X_{dB}$$

$$10 \times log_{10}\left(x^{m}\right) = m \times X_{dB}$$

Example: 2 is approximately 3 dB

$$8 = 2 \times 2 \times 2$$
 is approximately  $3 + 3 + 3 = 9$  dB



## The Decibel (dB)

To convert dB back to natural units, use this formula (this is for power, divide by 20 for voltage):

 $\left(x = 10^{\left(\frac{X_{dB}}{10}\right)}\right)$ 

$$10 \times log_{10}(x) = X_{dB}$$

$$log_{10}(x) = \frac{X_{dB}}{10}$$

$$10^{\log_{10}(x)} = x = 10^{\left(\frac{X_{dB}}{10}\right)}$$

Example: 20 dB

$$x = 10^{\left(\frac{20}{10}\right)} = 10^2 = 100$$

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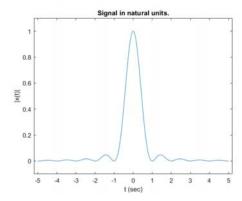


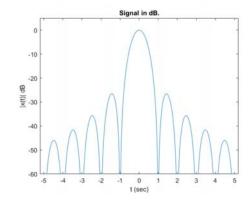
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# **Module 3.3 - Frequency Analysis Part 3**

## The Decibel (dB)

Plot where details when the signal is small are more evident in dB than natural form.





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#### **Student Exercise 1**

Two minutes.

The Fourier transform of a sinusoid is given below.

$$cos(2\pi f_0 t) \leftarrow \xrightarrow{Fourier} \frac{1}{2}\delta(f - f_0) + \frac{1}{2}\delta(f + f_0)$$

 $cos(2\pi f_0 t) \xleftarrow{Fourier} \frac{1}{2}\delta(f - f_0) + \frac{1}{2}\delta(f + f_0) \xrightarrow{\frac{1}{2}} f_0$ 

What will the DFT of this sinusoid look like?

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# Module 3.3 - Frequency Analysis Part 3

#### **FFT Resolutions**

Multiplication in the time domain is convolution in the frequency domain:

Time domain: element by element multiplication

Frequency domain: convolution

$$x_{w}[n] = x[n]w[n]$$

$$X_{w}[k] = X[k] \circledast W[k]$$

$$X_{w}(e^{j\omega}) = \frac{1}{2\pi} \int_{-\pi}^{\pi} X(e^{j\omega}) W(e^{j(\omega-\lambda)}) d\lambda$$

Matlab:  $xw = x \cdot w$ ;

 $W(e^{j\omega})$  the is the DTFT of w[n]



#### **FFT Resolutions**

Multiplication in the time domain is convolution in the frequency domain:

$$x_{w}[n] = x[n]w[n] \qquad X_{w}[k] = X[k] \circledast W[k]$$

$$x[n] \qquad \qquad \uparrow \qquad X[k]$$

$$w[n] \qquad \qquad W[k] dB$$

$$x_{w}[n] \qquad \qquad X_{w}[k] dB$$

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# **Module 3.3 - Frequency Analysis Part 3**

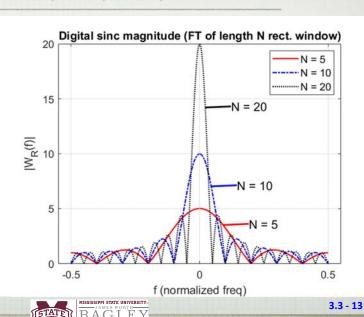
#### **FFT Resolutions**

The FFT of a rectangular window is called a *digital sinc*.

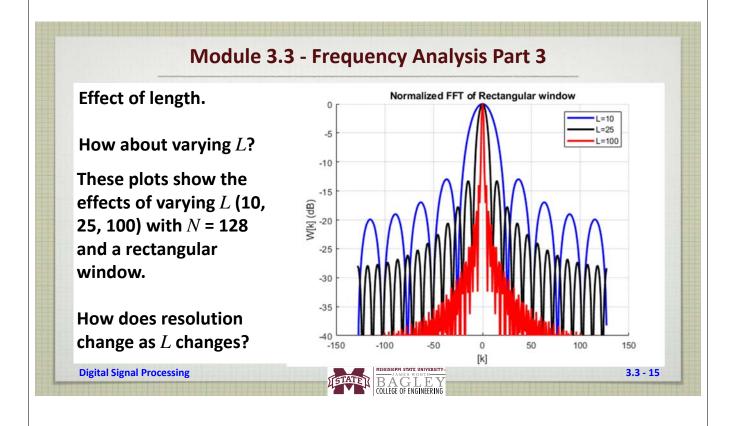
Please take time to review the DSP digital sinc handout and code.

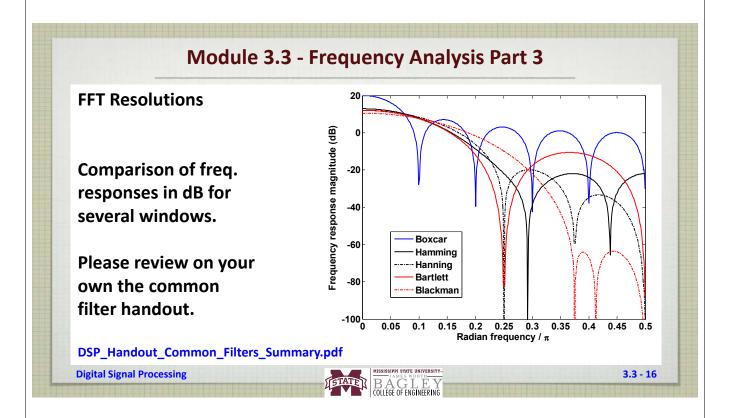
DSP\_Digital\_Sinc\_Example.m DSP\_Handout\_Digital\_Sinc.pdf

The real frequency is the normalized frequency times the sampling rate,  $f_S$ .



#### **Module 3.3 - Frequency Analysis Part 3** Normalized FFT **Student Exercise 2** Five minutes. Groups. -10 Post to discussion board. -20 Three different L=15 windows FFT responses in dB (normalized to 0 dB peak). -40 -50 What are differences? L=15Bartlett window -300 100 [k] **Digital Signal Processing** 3.3 - 14 BAGLEY COLLEGE OF ENGINEERING





# Module 3.3 - Frequency Analysis Part 3 Student Exercise 3

Thirty minutes. Instructor will demo experiment 1 and then you will work together in teams. We know that the computational FFT resolution is  $f_S/N$ . Lets do some experiments to see what the physical frequency resolution of the FFT is. To do this, we will run the following Matlab code and Excel files:

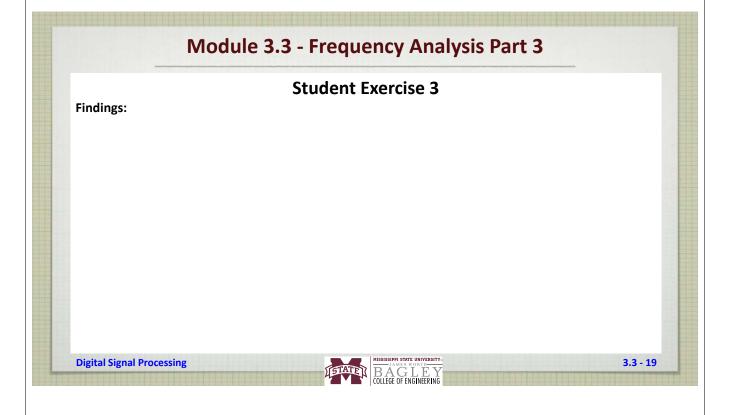
DSP\_FFT\_Resolution.m DSP\_FFT\_Resolution.fig

DSP\_FFT\_Resolution\_Group\_X.xlsx - these excel files are on OneDrive. Edit the one for your group.

Open Matlab. Open the file DSP\_FFT\_Resolution.m in the editor. Click on that file in your menu. At the top, click the Run icon.



#### **Module 3.3 - Frequency Analysis Part 3 Student Exercise 3** You should see this screen. DSP\_FFT\_Resolution DSP FFT Resolution GUI v. 1.0 The instructor will now Window Control Signal Control FFT Control demonstrate how to run Window Type the GUI for experiment 1 with Rectangular 64 Enable Sinudoid 1 100 The whole class. OHamming O Enable Sinudoid 2 Once you know, please complete the excel sheet provided for this exercise. The bottom two frequency plots are in dB to show results more clearly. The sampling rate is 500 Hz. 3.3 - 18 **Digital Signal Processing**



#### **FFT Resolution**

It can be shown (refer to Orfanidis reading materials) that the *FFT Physical Frequency Resolution* is given by

$$\Delta f_{PHYS} = C_{WIN} \frac{f_S}{L} = \frac{C_{WIN}}{LT_S} = \frac{C_{WIN}}{T_L}$$

where L is the window length in samples,  $f_S$  is the sampling rate in Hz (samples/second),  $T_S$  is the sampling interval (seconds/sample),  $T_L$  is the total observation time, that is,  $T_L = LT_S$ , and  $C_{WIN}$  is a constant depending on the window shape.  $C_{WIN} = \mathbf{1}$  for a rectangular and about 2 for a Hamming window.

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## **Module 3.3 - Frequency Analysis Part 3**

#### **FFT Resolution**

Thus the FFT has two resolutions:

Resolution	Equation	Controlled by
Physical	$\Delta f_{PHYS} = C_{WIN} \frac{f_S}{L}$	1. Sampling rate $(f_S)$ 2. Window type $(C_{WIN})$ 3. Window length $(L)$
Computational	$\Delta f_{COMP} = \frac{f_S}{N}$	<ol> <li>Sampling rate (f<sub>S</sub>)</li> <li>FFT Size (N)</li> </ol>

Note: A N-point FFT only analyzes N data points in the time domain. This means  $0 < L \le N$ .



# **Minute Paper**

Take one minute and fill out anonymous minute paper (on Canvas).

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# **Module 3.3 - Frequency Analysis Part 3**

# Appendix A

**Some Common Window Functions.** 

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