Eng. 100: Music Signal Processing DSP Lecture 2: Lab 2 overview

Curiosity: http://www.youtube.com/watch?v=qybUFnY7Y8w

Announcements:

- DSP lecture notes on Google Drive
- Lab 1

Finish RQ on Canvas by Thu. at 10:30AM (usually due *before* lab)
Finish Lab 1 this week; due Friday 5PM on Gradescope (grace)

- Read Lab 2 before next week's lab!
 Finish RQ on Canvas by next Thu. at 10:30AM.
- Midterm (on schedule): Wed. Mar. 20, in class
- See syllabus for office hours. Come say hi!

Outline

- Previous class summary (TC, A/D, frequency, Julia)
- Part 0. Lab 1 questions? (cut-and-paste vs understanding / HW0)
- Part 1: Terminology

Lab 2: Computing and visualizing the frequencies of musical tones

- Part 2. Sampling signals, especially sinusoids
- Part 3: Computing the frequency of a sampled sinusoid (with a computer, rather than by hand and eye like in previous class)
- Part 4: Visualizing, modeling and interpreting data using semi-log and log-log plots
- Part 5: Basic dimension analysis (units)

Part 1: Terminology

Terminology

Help me remember to define each new term! (First overview class may have been rushed but not now...)

```
Course title: Music Signal Processing What is a signal?
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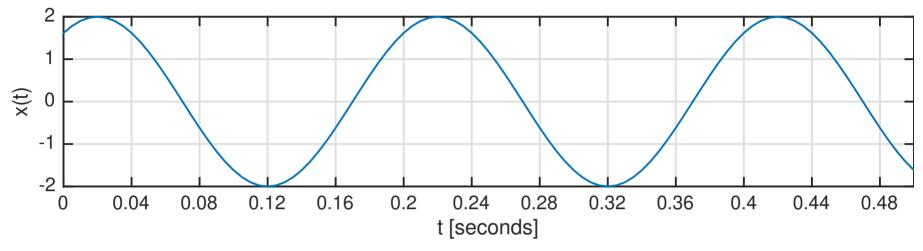
Wikipedia (electronics) 2012: a signal is any time-varying or spatial-varying quantity

2021: *"In signal processing, a signal is a function that conveys information about a phenomenon."*

Q0.1 Common use of "signal" ? (short answer; introduce neighbors)

Part 2: Sampling analog signals, especially sinusoidal signals

Sinusoidal "pure" tones (simple signal)



From Lab 1:

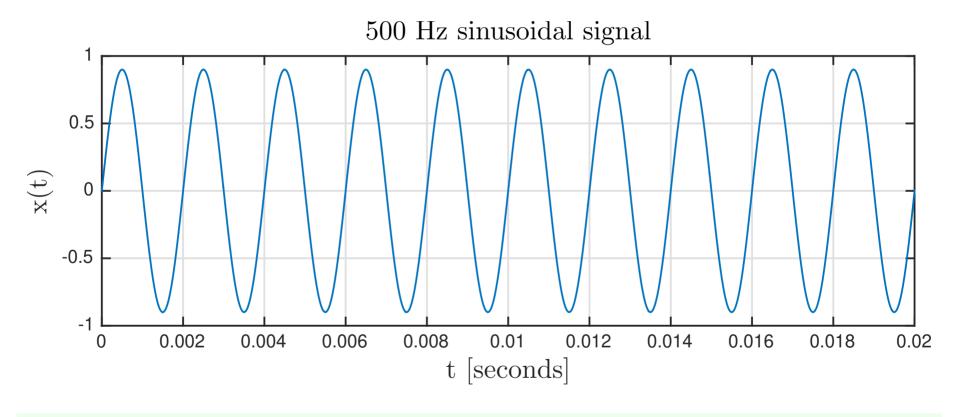
$$x(t) = 2\cos(2\pi 5 (t - 0.02)) = 2\cos(2\pi 5 t - \pi/5)$$

- Amplitude: A = 2
- Frequency: f = 5 Hz (cycles per second)
- Period: T = 1/f = 0.2 s
- Phase: $heta=-\pi/5$ radians

"musical?" (cf. instruments, cf. hearing range)

Sinusoidal signal at 500 Hz

$$x(t) = 0.9\cos(2\pi 500t - \pi/2)$$
 play

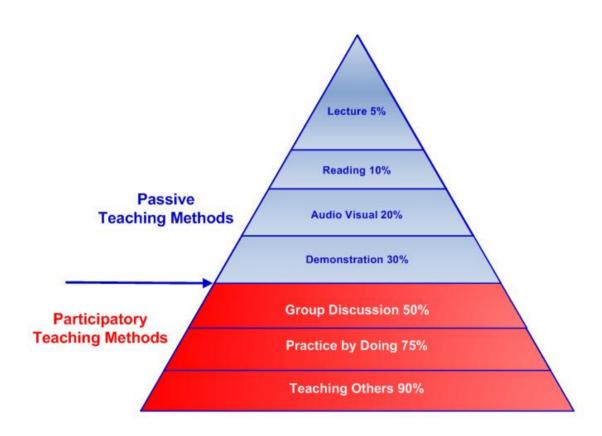


Q0.2 What is the period of this signal (in seconds)?A: 2B: 0.2C: 0.02D: 0.002E: 500

??

Learning pyramid

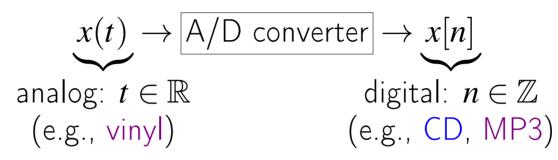
The Learning Pyramid



https://thepeakperformancecenter.com/educational-learning/learning/principles-of-learning/learning-pyramid (citation !!)

Sampling an analog signal

Analog signal (continuous-time signal): x(t), where t can be any real number. Units of "t" are seconds.



Crucial "system" quantities for an A/D converter (*i.e.*, for sampling):

• Sampling rate: S. Units: $\frac{\text{Sample}}{\text{Second}}$ or Hz

• Sampling interval:
$$\Delta = 1/S$$
. Units: seconds

On paper, or in software, sampling means: substitute t = n/S.

Digital signal (discrete-time signal):

$$x[n] = x(n/S) = x(n\Delta)$$

where *n* can be any *integer*.

Q0.3 What are the units of "n" above? (Choose best answer.) A: unitless B: seconds C: Hz D: $\frac{\text{Sample}}{\text{Second}}$ E: none of these ??

Q0.4 What are the units of "n/S" ? A: unitless B: seconds C: Hz D: $\frac{\text{Sample}}{\text{Second}}$ E: none of these ??

Analog signal x(t) and digital signal x[n] are related but quite different!

Terminology:

Q0.5 Common use of term *sample* or *sampling*?

Example: Sampled sinusoidal signal

Analog signal

$$x(t) = 2\cos(2\pi(5\mathrm{Hz})t - \pi/5)$$

Choose sampling rate: S = 50 Hz.

Q0.6 What is the sampling interval Δ ? A: 20 B: 2 C: 0.2 D: 0.02 E: None of these ??

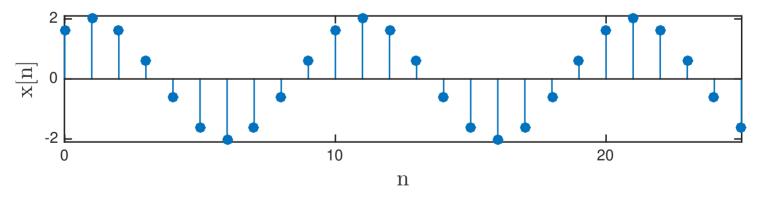
Mars Climate Orbiter loss 1999-08-23 ($\approx\$190$ M) https://science.nasa.gov/mission/mars-climate-orbiter Analog signal:

$$x(t) = 2\cos(2\pi(5\mathrm{Hz})t - \pi/5)$$

For S = 50 Hz, we substitute t = n/S = n/(50 Hz) to emulate A/D.

Digital signal:

$$\begin{aligned} x[n] &= x(0.02n) = 2\cos(2\pi(5\text{Hz})(n/50\text{Hz}) - \pi/5) \\ &= 2\cos(0.2\pi n - \pi/5) \text{ after simplifying} \end{aligned}$$



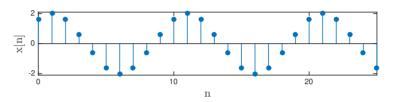
Q0.7 What are the units of " $0.2\pi n - \pi/5$ " above? (Choose best.) A: unitless B: radians C: Hz D: $\frac{\text{Sample}}{\text{Second}}$ E: degrees

??

Example: Sampled sinusoidal signal (continued)

Digital signal as a formula: $x[n] = 2\cos(0.2\pi n - \pi/5)$

Digital signal as a plot:



But in a computer (or DSP chip in phone) it is an array (list) of numbers:

$$\begin{array}{|c|c|c|c|c|} n & x[n] & x[n] \\ (value) & (formula) \\ \hline 0 & 1.62 & 2\cos(0.2\pi 0 - \pi/5) \\ 1 & 2.00 & 2\cos(0.2\pi 1 - \pi/5) \\ 2 & 1.62 & 2\cos(0.2\pi 2 - \pi/5) \\ 3 & 0.62 & \vdots \\ 4 & -0.62 & \vdots \\ 5 & -1.62 & \vdots \\ \vdots & \vdots & \end{array}$$

(Actually stored in binary (base 2) not in decimal.)

Sampling a sinusoid in general

Given a pure sinusoidal (analog) signal: $x(t) = A\cos(2\pi ft + \theta)$.

Q0.8 Units of the product ft? (not feet) A: unitless B: seconds C: Hz D: $\frac{\text{Sample}}{\text{Second}}$ E: none of these ??

If we sample it at $S \frac{\text{Sample}}{\text{Second}}$ (by substituting t = n/S), we get a digital sinusoidal signal (formula):

$$x[n] = A\cos(2\pi fn/S + \theta) = A\cos\left(2\pi \frac{f}{S}n + \theta\right)$$

Q0.9 Units of f/S? A: unitless B: seconds C: Hz D: $\frac{\text{Sample}}{\text{Second}}$ E: none of these 2?The quantity $\Omega = 2\pi f/S$ is called the *digital frequency* in DSP.

Sampling a sinusoid - summary

$$\begin{array}{c} \text{Analog} \\ \text{sinusoidal signal} \\ x(t) = A\cos(2\pi ft + \theta) \\ (\text{frequency } f) \end{array} \rightarrow \begin{array}{c} \text{Sampling} \\ \text{A/D} \\ \text{rate } S \end{array} \rightarrow \begin{array}{c} \text{Digital} \\ \text{sinusoidal signal} \\ \text{x}[n] = A\cos\left(2\pi \frac{f}{S}n + \theta\right) \\ (\text{list of numbers}) \end{array}$$

Real world:

- A computer sound card samples a microphone input signal.
- Applications like Shazam/Zoom/TikTok use digital audio signals.

Basic music transcription requires that we reverse this process!

$$\begin{array}{c} \text{Digital sinusoidal signal} \\ \times[n] \end{array} \rightarrow \begin{array}{c} \text{DSP} \\ \text{magic} \end{array} \rightarrow \begin{array}{c} \text{frequency (pitch)} \\ f \end{array}$$

Part 3: Computing the frequency of a sampled sinusoidal signal

Reconstructing a sinusoid from its samples

Given samples of a digital sinusoid in a computer:

$$x[n] = A\cos(\Omega n + \theta), \quad n = 1, 2, \dots, N.$$

(Stored as a list of N numbers, not as a formula, for known rate S.)

How can we find the frequency f of the original analog sinusoidal signal?

- Step 1. Determine the ''digital frequency'' $oldsymbol{\Omega}$
- Step 2. Relate Ω to the original (analog) frequency. (cf. "F = ma") This step is easy because $\Omega = 2\pi \frac{f}{S}$, so rearranging: $f = \frac{\Omega}{2\pi}S$.

Example. $x[n] = 3\cos(0.0632n + 5)$ with S = 8192 Hz. Original frequency is $f = \frac{0.0632}{2\pi} 8192$ Hz = 82.4Hz (low E)

But what if we are given an array of signal values instead of a formula?

Finding a digital frequency

(The computer or DSP chip perspective)

Given:

• Signal values:

 $|x|n| = A\cos(\Omega n + \theta)$

- Sinusoidal assumption (model):
- Goal: Determine the digital frequency Ω , a model parameter.

This problem arises in many applications, including music DSP. EE types have proposed many solutions. Lab 2 uses an elegantly simple method based on trigonometry.

Everything in signal processing / data science / statistics starts with arrays of numbers (data) and uses models to extract (hopefully useful) information.

Key trigonometric identities

Angle sum and difference formulas [wiki]

$$cos(a+b) = cos a cos b - sin a sin b$$

$$cos(a-b) = cos a cos b + sin a sin b$$

Product-to-sum identities:

$$2\cos a \cos b = \cos(a-b) + \cos(a+b)$$

$$2\sin a \sin b = \cos(a-b) - \cos(a+b)$$

Wikipedia says these identities date from 10th century Persia. [wiki]

We will use these (ancient) identities repeatedly!

Practical example: Tuning a piano

Rewriting product-to-sum identity:

$$\cos(a+b) + \cos(a-b) = 2\cos a \cos b.$$

Substitute $a = 2\pi 442t$ and $b = 2\pi 2t$:

 $\cos(2\pi 444t) + \cos(2\pi 440t) = 2\cos(2\pi 442t)\cos(2\pi 2t).$

The sum of two sinusoids having close frequencies is a sinusoid at the average of the frequencies with a (sinusoidally) time-varying amplitude!

The combined (sum) signal gets louder and softer, "beating" with period = 0.25 sec. Why do we consider the sum? ?? The slower the period, the closer the 2 frequencies.

Why is this relevant to tuning a piano (or guitar or ...)? $\boxed{??}$

Back to finding a digital frequency

Repeating product-to-sum identity:

$$\cos(a+b) + \cos(a-b) = 2\cos a \cos b.$$

As another practical application of this identity, some clever DSP expert suggested substituting $a = \Omega n + \theta$ and $b = \Omega$:

$$\cos(\Omega(n+1)+\theta)+\cos(\Omega(n-1)+\theta)=2\cos(\Omega)\cos(\Omega n+\theta).$$

Now use sinusoidal model assumption: $x[n] = A\cos(\Omega n + \theta)$, yielding: $x[n+1] + x[n-1] = 2\cos(\Omega)x[n]$. Rearranging yields $\cos(\Omega) = \frac{x[n+1] + x[n-1]}{2x[n]}$, or equivalently:

$$\Omega = \arccos\left(\frac{x[n+1] + x[n-1]}{2x[n]}\right).$$

Example: find a sinusoid's digital frequency

A: 1B: 2C: 3D: 4E: 5Earlier example:n = 012345...x[n]1.622.001.620.62-0.62-1.62...

Apply the arccos formula to this data using n = 2:

Q0.10 How many signal samples do we need to find Ω ?

$$\Omega = \arccos\left(\frac{x[n+1] + x[n-1]}{2x[n]}\right) = \arccos\left(\frac{x[3] + x[1]}{2x[2]}\right)$$
$$= \arccos\left(\frac{0.62 + 2}{2 \cdot 1.62}\right) = \arccos(0.8086) = 0.629 \approx 0.2\pi.$$

cf. sampled sinusoidal signal on p. 13.

Is this useful for a guitar tuner app yet? [??]

??

Example of finding a sinusoid's frequency

Recall from ''Step 2'' on p. 17 that
$$f=rac{\Omega}{2\pi}S$$

Combining Step 1 and Step 2:

$$f = \frac{S}{2\pi} \arccos\left(\frac{x[n+1] + x[n-1]}{2x[n]}\right).$$

Use this formula in Lab 2 to compute the frequency from a digital signal corresponding to samples of a sinusoid.

Example: S = 1500 Hz and x[n] = (...,?,?,?,3,7,4,?,?,?,...)The signal values denoted "?" are, say, lost or garbled. Solution: $f = \frac{1500}{2\pi} \arccos\left(\frac{3+4}{2\cdot7}\right) = \frac{1500}{2\pi} \arccos\left(\frac{1}{2}\right) = \frac{1500\pi}{2\pi} = 250$ Hz.

Historical note: this approach is a simplification of Prony's method from (!) 1795.

Exercise

Suppose S = 8000 Hz and the signal samples are x[n] = (..., 0, 5, 0, -5, 0, 5, 0, -5, 0, ...)

Q0.11 What is the frequency f of the sinusoid (in Hz)? A: 1000 B: 2000 C: 4000 D: 8000 E: None of these

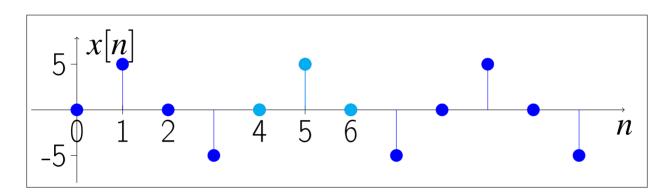
Hint. Here is the arccos frequency formula repeated for convenience:

$$f = \frac{S}{2\pi} \arccos\left(\frac{x[n+1] + x[n-1]}{2x[n]}\right).$$

??

Illustration

Stem plot of the signa x[n] from previous page.



(Dis)Advantages of this method

Many frequency estimation methods have been proposed. Apparently not everyone just uses this one; why not?

Advantages of this method:

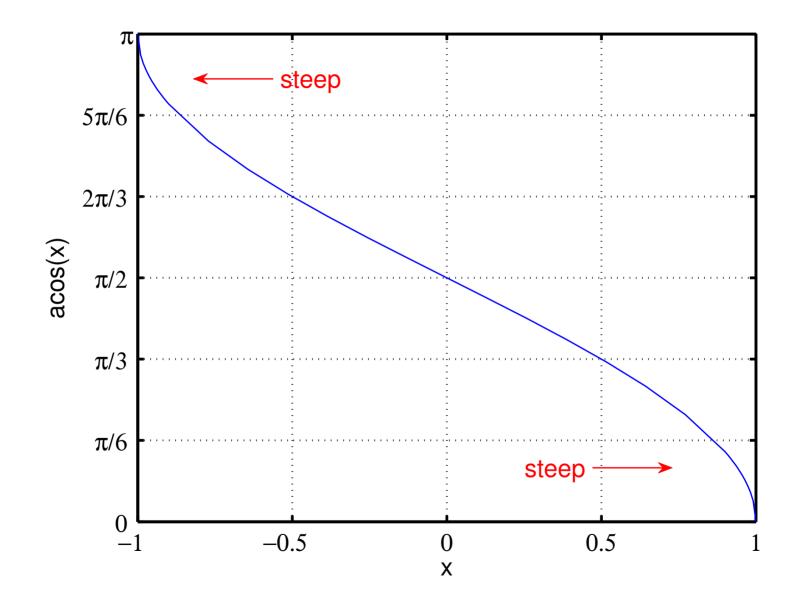
- Very simple to implement, can use simple DSP chip.
- Fast tracking of sudden frequency changes.
- Can use outliers (weird values) to segment long signals.
- Requires knowledge of trigonometry only.

Disadvantages of this method

- Very sensitive to additive noise in the data x[n].
- What if x[n] = 0 for some n? Divide by 0!
- Arc-cosine function is sensitive to small changes.

A useful starting point for music DSP...

Arc-cosine



Part 4: Visualizing, modeling, and interpreting data using semi-log and log-log plots

Data visualization and modeling

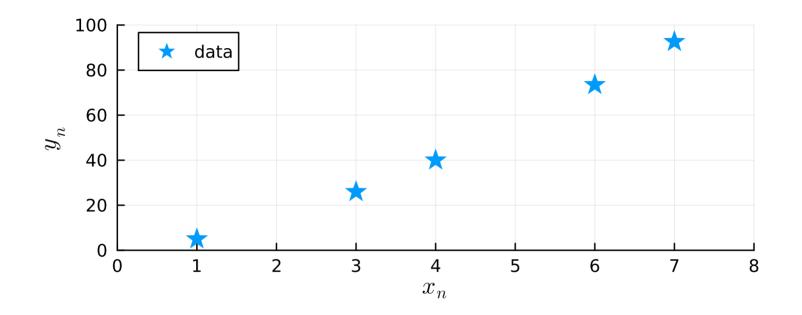
Given: N pairs of data values: $(x_1, y_1), (x_2, y_2), \dots, (x_N, y_N)$

Goal: Find a relationship between the values (a model): $\circ y = g(x)$ $\circ y_n = g(x_n), n = 1, \dots, N$

- Example (Physics 140):
 x = height above ground a ball is released
 y = velocity on impact with ground.
- Example (Physics 240):
 x = electrical current through a light bulb
 y = energy released in the form of heat by the bulb
 - Example (Engin 100-430):
 - x = piano key "number" (1 to 88)
 - y = frequency of note played by a key

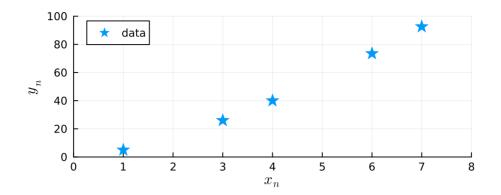
Visualizing data using scatter plots

Example (scatter plot of 5 pairs of data values):



Do your eyes try to "connect the dots?" Your brain is trying to build a model!

Making a scatter plot in Julia



```
using Plots; default(markerstrokecolor=:auto, markersize=8)
using LaTeXStrings # optional: for plot labels with equations
x = [1, 3, 4, 6, 7]
y = [5.00, 25.98, 40.00, 73.48, 92.60]
scatter(x, y; marker=:star, label="data", size=(500,200),
xaxis=(L"x_n", (0,8), 0:8), yaxis=(L"y_n", (0,100)))
# savefig("scatter1.pdf")
```

Julia notes:

• The arguments x, y here are called "positional arguments" and their order matters.

- The arguments xlabel=L"x_n" etc. are called named keyword arguments or named parameters or name-value pairs in Julia and Python. (Matlab has some such capability in recent versions.) They are optional and they can appear in any order (after the semicolon).
- The :star value here is a **Symbol** based on string interning.

Common mathematical models

=ax

Linear model: one parameter: slope *a*

Affine model:
$$y = ax + b$$

two parameters: slope *a* and intercept *b*

Quadratic (parabola) model:
$$y =$$

$$y = ax^2 + bx + c$$

Simple "power" model: power parameter p, scale factor b

Simple "exponential" model: $y = b a^x$ Note that the independent variable x is in the exponent here. Q0.12 How many parameters does the quadratic model have?A: 0B: 1C: 2D: 3E: 4

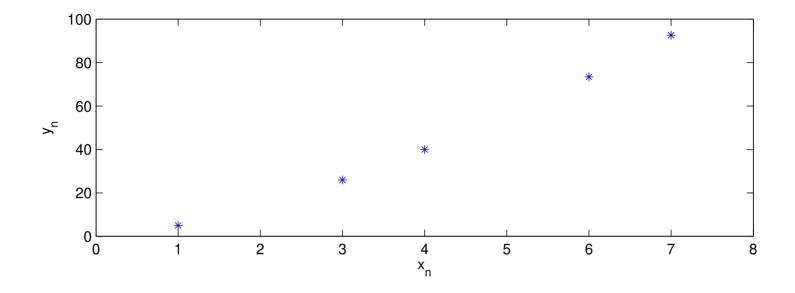
Q0.13 Which model is appropriate for height/velocity example? (skip: Phys 140) A: linear B: affine C: quadratic D: power E: exponential

How do we choose among these models given data?

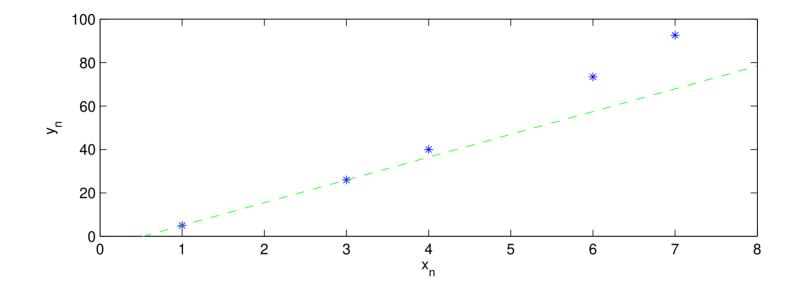
??

??

Example: Linear or affine model?



Example: Linear or affine model?



Any two points determine the equation for a line. Does a line fit this data?

So we rule out both linear and affine models by visualization.

For power and exponential model, we need logarithms.

Review of logarithms

- In Julia, C, C++, python, Matlab and in this class, log means natural log, (base e), which might be ln on your calculator.
- For base-10 logarithm use log10 in Julia, C, C++, Matlab, python, ..., and write log10 on paper.
- Properties of logarithms (for any base b > 0): (Math 105)
 - $b^{\log_b(x)} = x$ if x > 0 (the defining property)
 - $\log_b(xy) = \log_b(x) + \log_b(y)$ if x > 0 and y > 0 log of

•
$$\log_b(x^p) = p \log_b(x)$$
 if $x > 0$

• Other related properties:

• $e^{\log(x)} = x$ and $10^{\log_{10}(x)} = x$ if x > 0

•
$$\log(e) = 1$$
 and $\log_{10}(10) = 1$ and $\log_b(1) = 0$

•
$$e^{a+b} = e^a e^b$$

Q0.14Exercise: Simplify $e^{c \log(z)}$ for z > 0.A: z^c B: c^z C: ze^c D: czE: None of these

Simple "exponential" model

$$y = b a^x$$

Take the logarithm (any base) of both sides:

$$\log(y) = \log(b a^x) = \log(b) + \log(a^x) = \log(b) + x \log(a)$$
$$\log(y) = \log(a) x + \log(b)$$

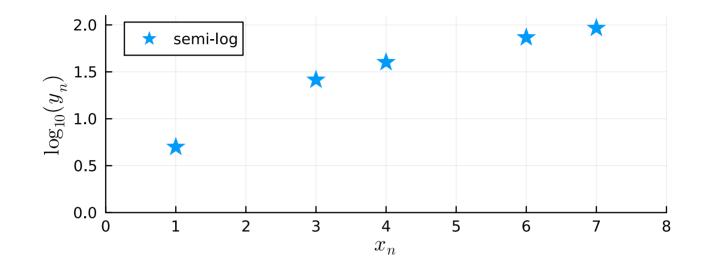
This is the equation of a line on a log scale:

$$\underbrace{\log(y)}_{\tilde{y}} = \underbrace{\log(a)}_{\text{slope}} x + \underbrace{\log(b)}_{\text{intercept}}$$

To see if the exponential model fits some data, make a scatter plot of $\log(y_n)$ versus x_n and see if it looks like a straight line.

This is called a semi-log plot.

Making a semi-log plot in Julia



```
using Plots; default(markerstrokecolor=:auto, markersize=8)
using LaTeXStrings
x = [1, 3, 4, 6, 7]
y = [5.00, 25.98, 40.00, 73.48, 92.60]
scatter(x, log10.(y); marker=:star, size=(500,200), label="semi-log",
xaxis=(L"x_n", (0,8), 0:8), yaxis=(L"\log_{10}(y_n)", (0,2.1)))
# savefig("scatter2.pdf")
```

Q0.15 The <mark>exponential model</mark> provides a good fit for this data. A: True B: False

Simple "power" model

$$y = b x^p$$

Take the logarithm (any base) of both sides:

$$\log(y) = \log(bx^p) = \log(b) + \log(x^p) = \log(b) + p\log(x)$$
$$\log(y) = p\log(x) + \log(b)$$

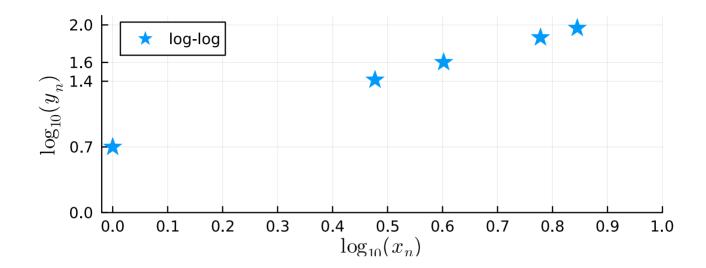
This is the equation of a line on a log-log scale:

$$\underbrace{\log(y)}_{\tilde{y}} = \underbrace{p}_{\text{slope}} \underbrace{\log(x)}_{\tilde{x}} + \underbrace{\log(b)}_{\text{intercept}}$$

To see if the power model fits some data, make a scatter plot of $log(y_n)$ versus $log(x_n)$ and see if it looks like a straight line.

What do you suppose this type of plot is called? $\hfill \ref{eq:constraint}$

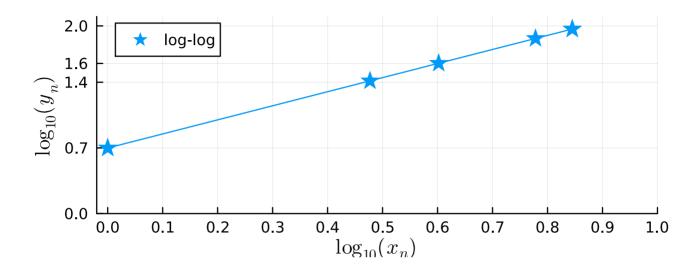
Making a log-log plot in Julia



using Plots; default(markerstrokecolor=:auto, markersize=8)
using LaTeXStrings
x = [1, 3, 4, 6, 7]
y = [5.00, 25.98, 40.00, 73.48, 92.60]
scatter(log10.(x), log10.(y); marker=:star, size=(500,200), label="log-log",
xaxis=(L"\log_{10}(x_n)", (-0.02,1), 0:0.1:1),
yaxis=(L"\log_{10}(y_n)", (0.,2.1), [0, 0.7, 1.4, 1.6, 2]),)
savefig("scatter3.pdf")

Q0.16 The power model provides a good fit for this data. A: True B: False





scatter!(log10.(x), log10.(y); smooth=true) # along with other style args...

smooth=true : adds "least squares best-fit line" to scatter plot

Yes! log-log plot lies along a line \implies power model is a good fit. Now we just need the parameters to write our equation model. \circ intercept $\approx 0.7 = \log_{10}(b) \implies b = 10^{0.7} \approx 5.0$ \circ slope $= \frac{\Delta \tilde{y}}{\Delta \tilde{x}} \approx \frac{1.6-0.7}{0.6-0.0} = 1.5 \implies p = 1.5$ Model: $y = b x^p = 5 x^{1.5}$

Checking a model

Recall given data:

```
x = [1, 3, 4, 6, 7]
y = [5.00, 25.98, 40.00, 73.48, 92.60]
```

Model found on previous slide: $y = 5x^{1.5}$

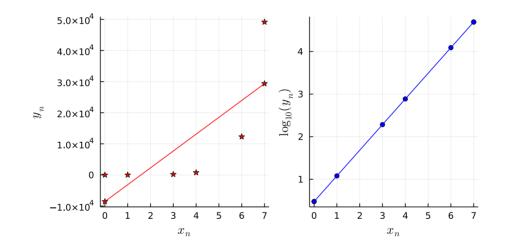
To check model in Julia (using **broadcast**): 5 * (x .^ 1.5)

Output is: 5-element Vector{Float64}: 5.0 25.981 40.0 73.485 92.601

So our power model fits the given data very well.

A semi-log example

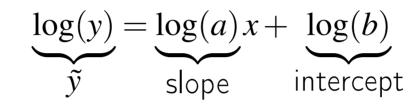
```
using Plots; default(label="")
x = [0, 1, 3, 4, 6, 7]
y = [3, 12, 192, 768, 12288, 49152]
p1 = scatter(x, y; color=:red, marker=:star, smooth=true)
p2 = scatter(x, log10.(y); color=:blue, marker=:circle, smooth=true)
plot(p1, p2)
```



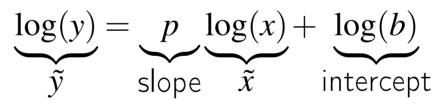
Q0.17 Exercise: determine a model for this data. (after Lab 2 overview)

Summary of two important models

• Simple exponential model: $y = b a^x$ Use semi-log plot:



• Simple power model: $y = b x^p$ Use log-log plot:

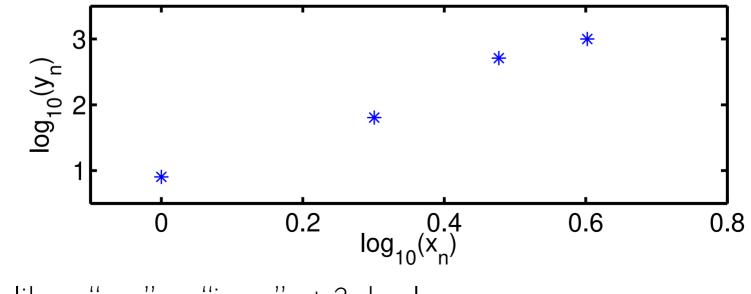


Even though Lab 2 uses these models for musical notes, they are ubiquitous in science and engineering and more.

Models with missing data (read yourself)

Musical context: song without all 88 notes

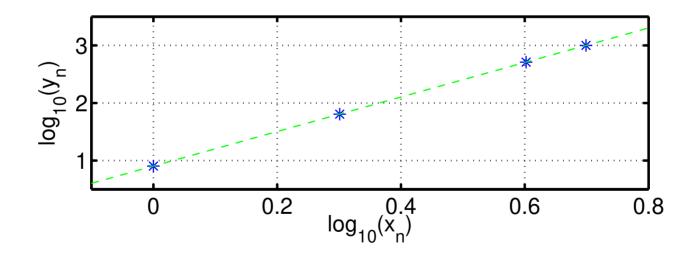
First try: x = 1:4; scatter(log10.(x), log10.(y))



Looks like a "gap" or "jump" at 3rd value

Example with missing data continued

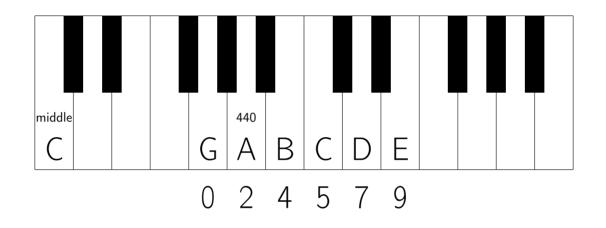
Second try: using Plots y = [8, 64, 512, 1000] x = [1, 2, 4, 5] scatter(log10.(x), log10.(y); smooth=true)



slope =
$$(3-0.9)/(0.7-0) = 3 = p$$

intercept = $0.9 = \log_{10}(b) \implies b = 10^{0.9} = 8$
Model: $y = 8x^3$

Missing frequencies in "The Victors"



Missing: 1 3 6 8 10 11 12 13 ... -1 -2 -3 ...

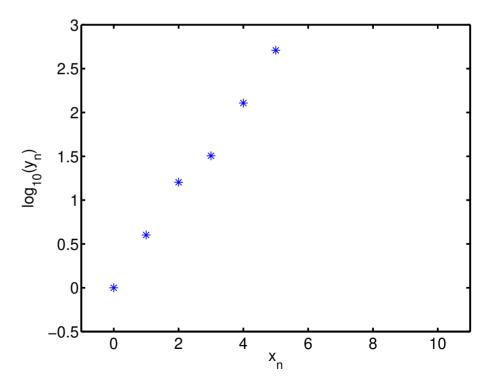
Lab 2 has lots of missing data!

"The Victors" only uses a few of the 88 keys on a piano.

Example: y = [1, 4, 16, 32, 128, 512]where each x_n is one of the numbers in the set $\{0, 2, ..., 87\}$

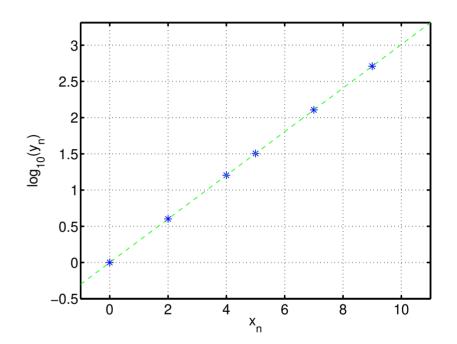
First try: x = 0:5 scatter(x, log10.(y))

Lots of jumps!



Lots of missing data continued

Second try: y = [1, 4, 16, 32, 128, 512] x = [0, 2, 4, 5, 7, 9] scatter(x, log10.(y))



slope = $(2.7 - 0)/(9 - 0) = 0.3 = \log_{10}(a) \implies a = 10^{0.3} = 2$ intercept = $0 = \log_{10}(b) \implies b = 10^0 = 1$

Model: $y = 2^x$

You will see a similar situation in Lab 2 (with different data).

What you will do in Lab 2

- Download a sampled signal from Canvas site: A tonal version of the chorus of "The Victors."
- Load into Julia; segment (chop up) into notes using reshape.
- Apply arccos formula to compute frequency of each note.
- Make log-log and semi-log plots of frequencies.
- Determine the formula relating frequencies of notes.
- Note: "Accidentals" are all missing; but you can infer their existence & frequencies from your plot!

Part 5: Basic dimension analysis by example (read yourself if time runs out in class)

Dimensional Analysis Example 1

- Goal: Determine formula for the period of a swinging pendulum, without any physics!
- Find ingredients: mass, length, gravity
- Model: Period = $(mass)^a$ $(length)^b$ g^c where g = acceleration of gravity 9.80665m/s^2 a, b, c are unknown constants to be found
- Approach: Find exponents using dimensional analysis: time = $(mass)^a$ $(length)^b$ $(length/time^2)^c$

• No "mass" on LHS so
$$a = 0$$

• No "length" on LHS so $0 = b + c$
• For time: $1 = -2c \Longrightarrow c = -1/2$, so $b = 1/2$

Model: period = length^{1/2} $g^{-1/2}$ = (length /g)^{1/2} From physics: period = 2π (length /g)^{1/2}

 $(2\pi$ is a unitless constant; cannot be found from dimension analysis.)

Dimensional Analysis Example 2

Prof. Yagle asks:

- If 1.5 people can build 1.5 cars in 1.5 days, how many cars can 9 people build in 9 days?
- Do problems like this give you a headache?
- Would you like to solve problems like this with minimal thinking?

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Given: (1.5 cars) / (1.5 days) / (1.5 people) = 2/3 cars / day / people

Now match units: (2/3 cars / day / people) (9 days) (9 people) = 54 cars

Simply matching the units suffices.

Summary

- Sampling: a computer can determine frequency of a pure sinusoid from 3 consecutive samples.
- Semi-log plot of $y = b a^x$ (exponential model) is a straight line.
- Log-log plot of $y = bx^p$ (power model) is a straight line.
- Dimensional analysis often gives you correct answers.

Assignment: Read Lab 2 before (next week's) lab!

References

[1] M-Z. Poh, N. C. Swenson, and R. W. Picard. A wearable sensor for unobtrusive, long-term assessment of electrodermal activity. *IEEE Trans. Biomed. Engin.*, 57(5):1243–52, May 2010.