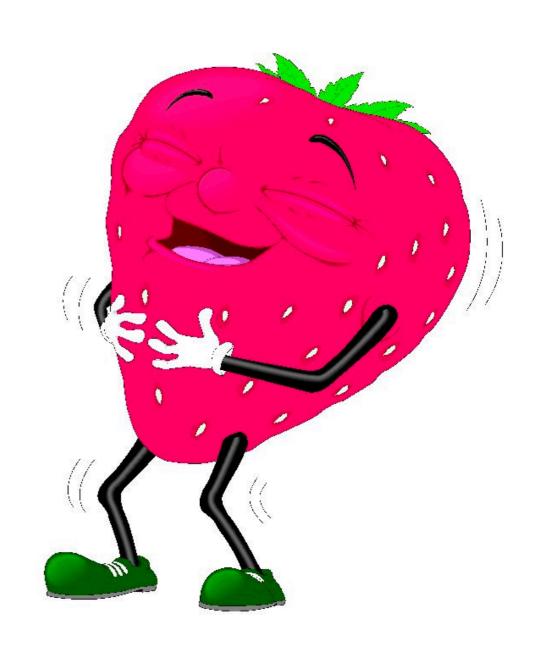
## TIMESERIES 4

10.28.2020

### PROBLEM SET 4

\* is due Friday!



#### PROBLEM SET 5

- \* will be posted Friday
- \* but you will have 3 weeks to finish it

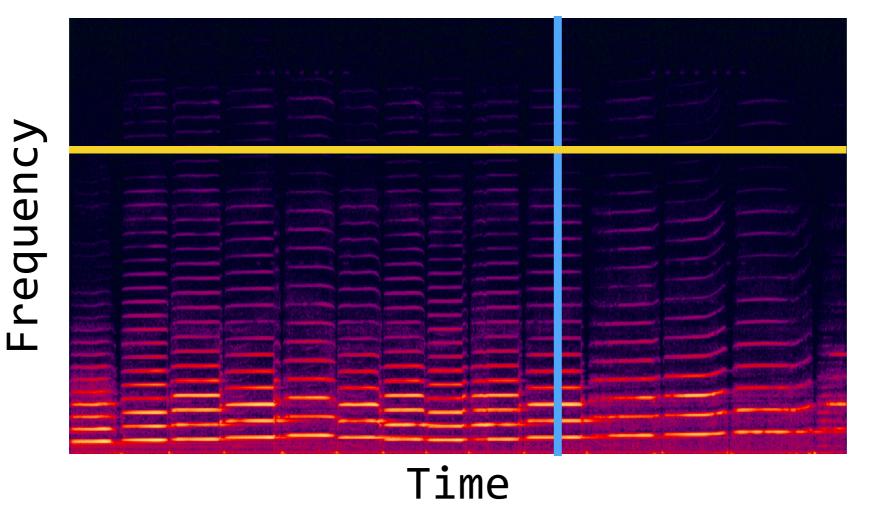


#### RECAP: SPECTROGRAM

- \* if we compute the fourier transform for small snippets of time and then stack them together into an array
  - \* this is the **spectrogram**
  - \* it shows which frequencies are present in a timeseries at each point in time
  - \* you should know how to read a spectrogram

#### THE SPECTROGRAM

- \* each column is the fourier transform of a short snippet
- \* what about each row? what does one row mean?



- \* **filtering** is a process that removes some frequencies from a timeseries and lets others remain (or even amplifies them)
- \* this is accomplished by convolving your timeseries with a **filter**, a small array that is designed to have a specific effect

- \* low-pass filter: removes high frequencies, allows low frequencies through
- \* high-pass filter: removes low frequencies, allows high frequencies through
- \* band-pass filter: removes all frequencies except for a specific band (the "pass band")

- \* low-pass filter: removes high frequencies, allows low frequencies through
- \* high-pass filter: removes low frequencies, allows high frequencies through
- \* band-pass filter: removes all frequencies except for a specific band (the "pass band")

- \* back to the spectrogram:
  - \* one row of a spectrogram is a lot like a band-pass filtered version of a timeseries

- \* suppose we have some EEG data from a human subject and we want to filter it so that only alpha-band oscillations remain
  - \* (this is a band-pass filter)
- \* how do you make a filter that has the properties you want?

- \* **scipy.signal** is a module in scipy that contains lots of useful functions for filter design
- \* scipy.signal.firwin creates "finite impulse response" filters with desired properties

#### ANALYZING A FILTER

- \* scipy.signal.freqz is a great function that tells you what the *frequency* response of your filter looks like
  - \* i.e. it tells you what the filter is going to do to your signal

# RECALL: FOURIER ANALYSIS

- \* fourier transforms have an interesting property related to convolution:
- \* given two timeseries, f and g, the fourier transform of their convolution = the element-wise product of their fourier transforms

$$FT(f*g) = F \cdot G$$

\* the reverse is also true:

$$F * G = FT(f \cdot g)$$

#### Time Function Sinc Boxcar $-\frac{\tau}{2}$ 0 $\frac{\tau}{2}$ $G(t) = \begin{cases} 1-|t|/\tau, |t| < \tau \\ 0, |t| > \tau \end{cases}$ Sinc<sup>2</sup> Triangle 0 $G(t) = e^{-1/2t^2}$ Gaussian Gaussian τ DC Shift Impulse $G(t) = \delta(t)$ = 0, t ≠ 0 0 Single Freq. $G(t) = \cos \omega_0 t$ Sinusoid π/ω 0 π/ω 2π/ω G(t) = comb(t)Comb. Comb. $= \sum_{n=0}^{\infty} \delta(1-n\tau)$ 0 τ 2τ 3τ -2π

Frequency Function

 $S(f) = \tau \operatorname{sinc}(f\tau)$ 

 $S(f) = \tau \operatorname{sinc}^2(ft)$ 

 $= (1/\pi f) \sin (\pi f t)$ 

 $= (1/\pi^2 f^2 \tau) \sin^2 (\pi f t)$ 

-1/<sub>\tau</sub> 0 1/<sub>\tau</sub> 2/<sub>\tau</sub> 3/<sub>\tau</sub> 4/<sub>\tau</sub>

 $S(f) = \frac{1}{2}(\delta(f+f_0) + \delta(f-f_0))$ 

2π

 $4\pi$ 

 $S(f) = \tau(2\pi)^{1/2} e^{-(\pi f \tau)^2}$ 

-1/<sub>T</sub> 0 1/<sub>T</sub>

0

0

 $S(f) = \sum_{-\infty}^{\infty} \delta(f-n/\tau)$ 

S(f) = 1

 $-f_0$ 

# END