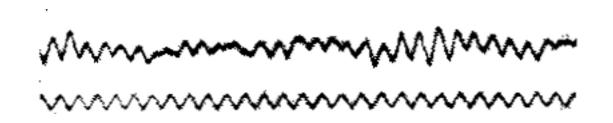
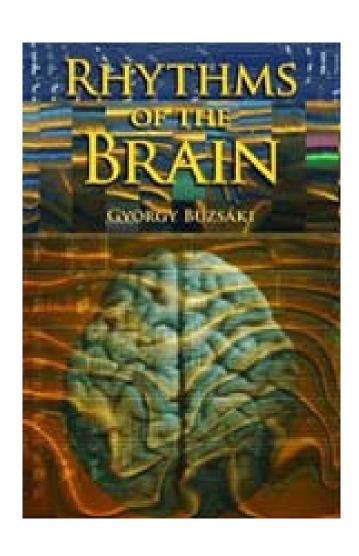
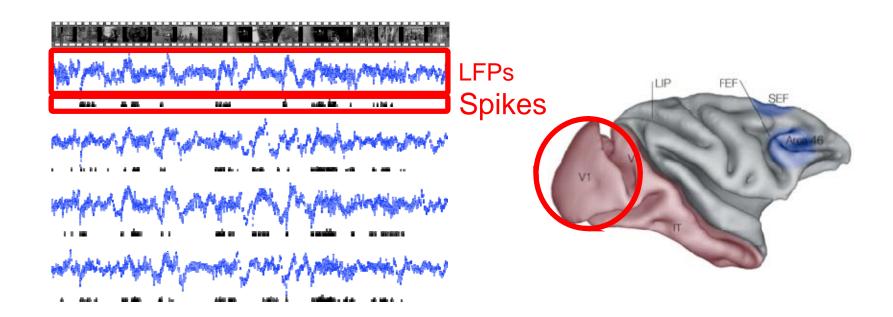
Computational Neuroscience [Tutorial]

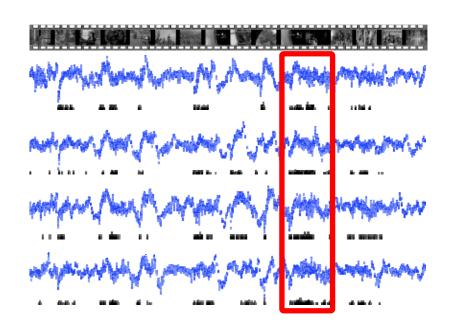


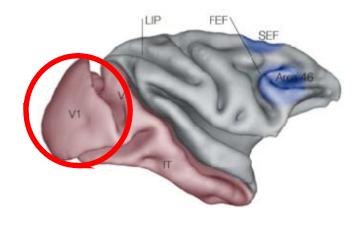
- "Oscillations" is an hot topic in neuroscience
- The importance of the role played by oscillations has been made clear by several studies
- However the exact role of oscillations in stimulus coding is still not well understood: which are the important rhythms? Which neural phenomenon corresponds to which frequency?

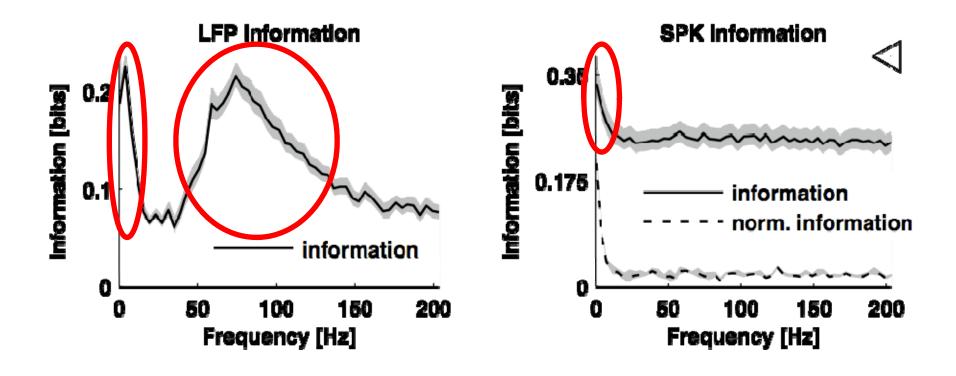


- In this tutorial we will focus on the analysis presented in *Belitski et al. Journal of Neuroscience 2009*
- The aim of this work was to understand which are the most important rhythm in the cortex from a decoding point of view
- The goal is to provide the fundamental tools for to reproducing the same type of analysis on your own / on your data







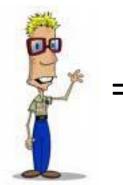


Organization of the tutorial

- PART I Introduction to Fourier-based spectral analysis techniques
- PART II Introduction to the bias problem and to the information breakdown toolbox
- PART III: how to perform the single-frequency information analysis

During this tutorial we'll try to...

- Use as little math as possible
- Give a look to some simple Matlab example scripts



= thing are getting technical !!!

PART I – INTRODUCTION TO FOURIER SPECTRAL ANALYSIS TECHNIQUES

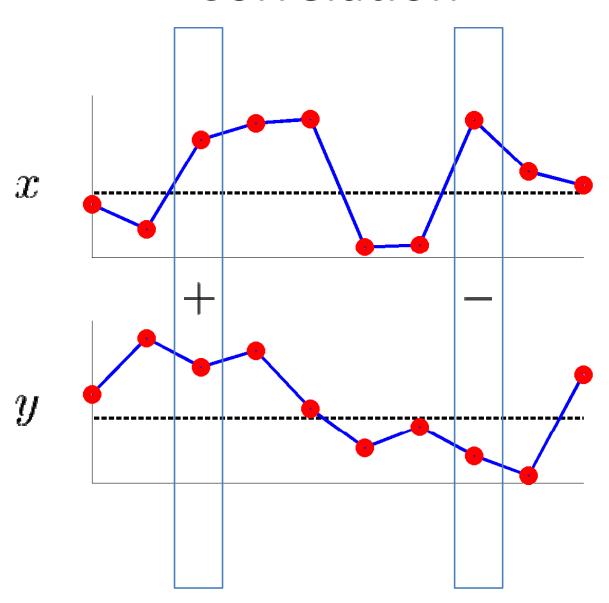
The Underpinnings of the Fourier Transform

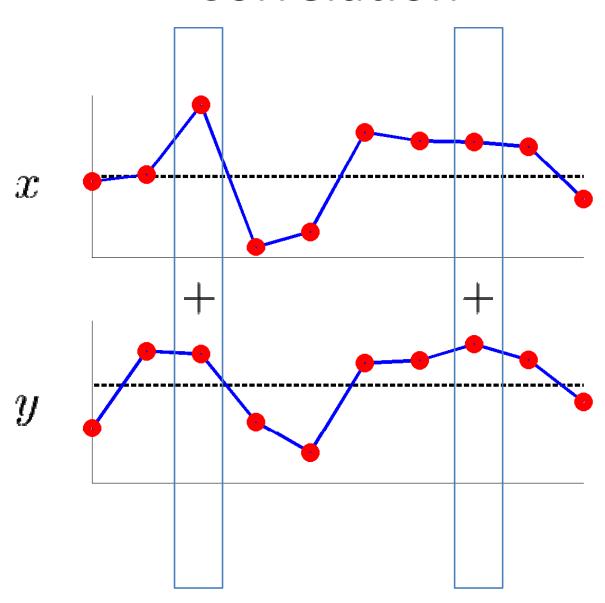
• THEOREM:

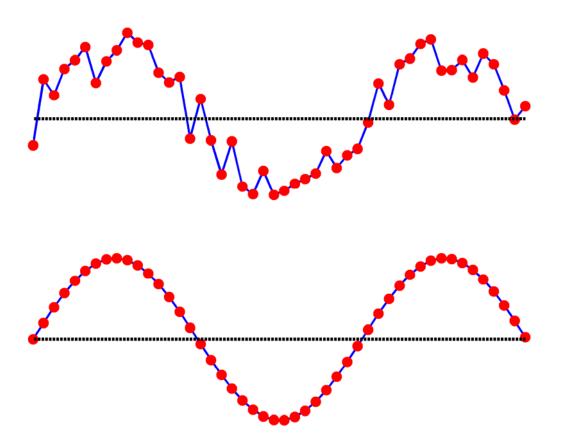
- any time-series can be expressed as a discrete weighted sum of oscillatory functions: sines and cosines
- If the time-series is N-points long then we only need to consider N/2 distinct frequencies
- These frequencies are equally-spaced in the range
 OHz to Fs/2, Fs being the sampling frequency

- How do we extract the coefficient for each sine and cosine?
- In general, how can we measure the similarity between two (zero-mean) time series?

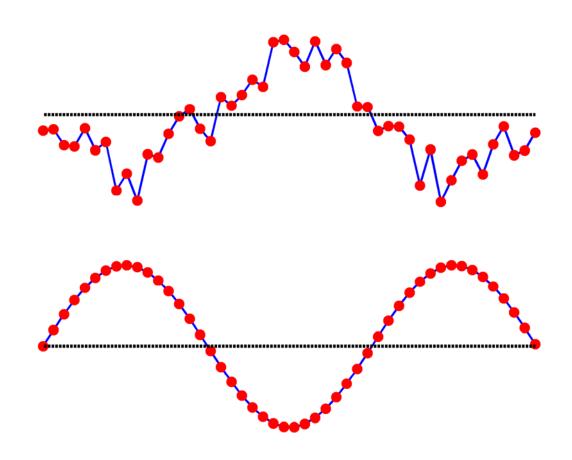
$$\frac{1}{N} \sum_{i=1}^{N} x[i] \cdot y[i]$$

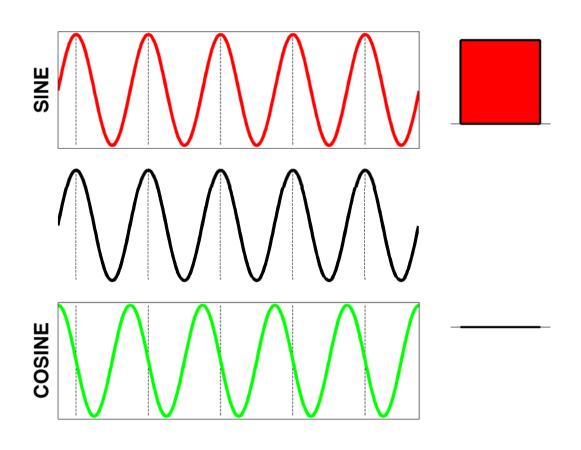


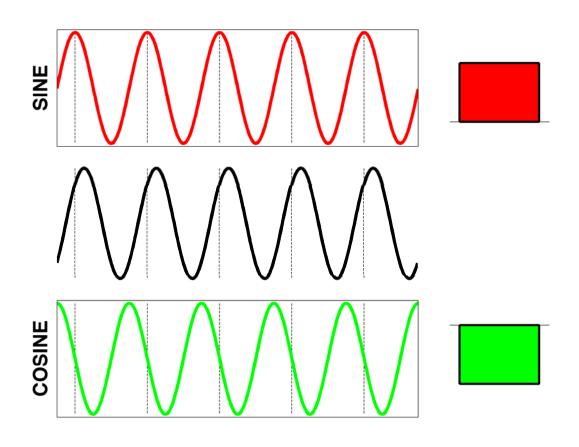


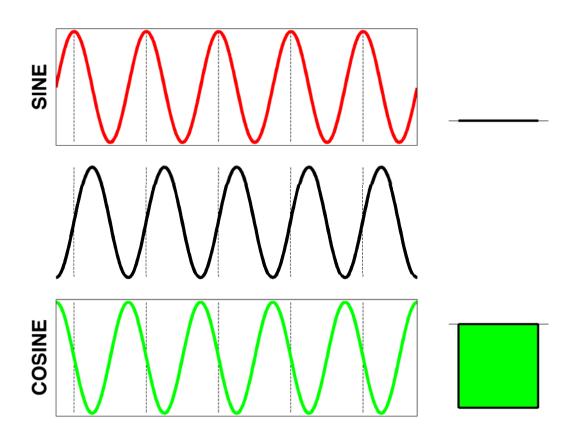


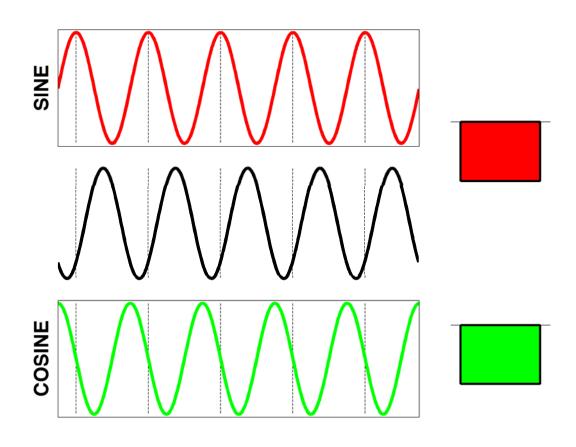
But what if we are in THIS situation?

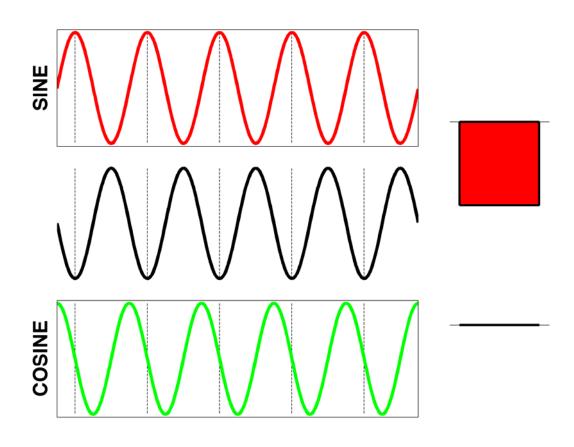


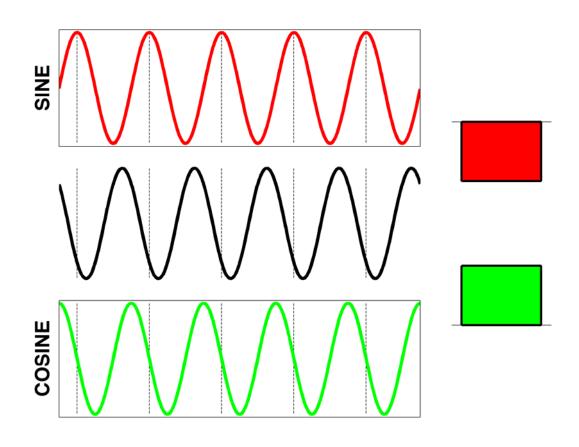


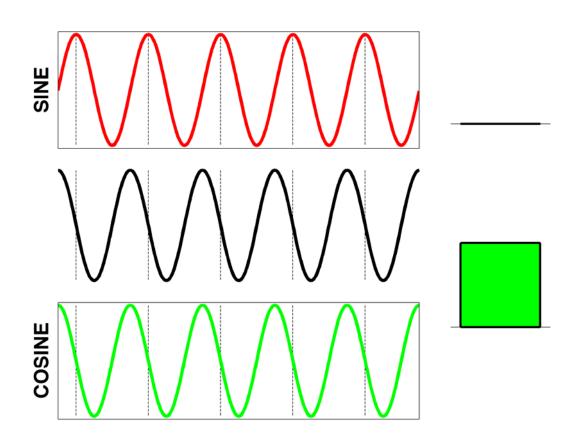


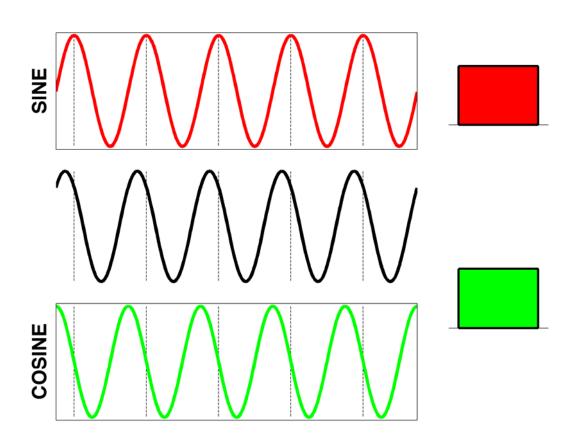


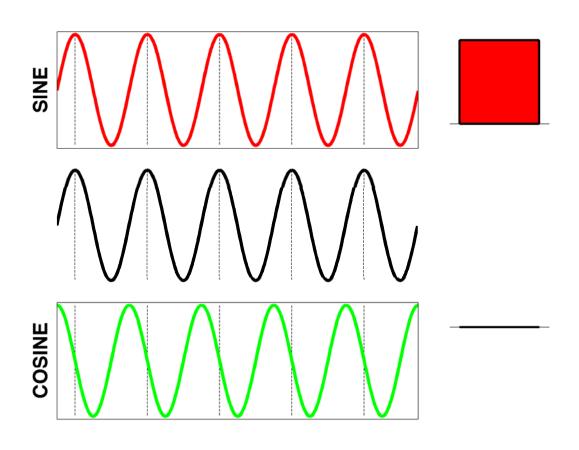






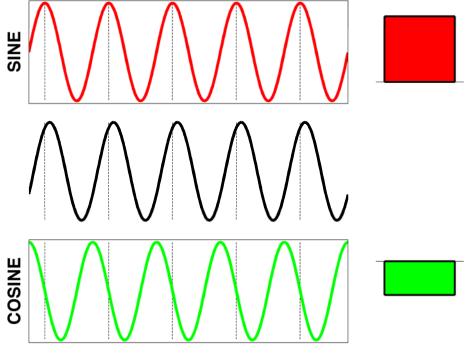






• **SOLUTION:** for a given frequency *f*, we simply need to compute the correlation between our time series and both a sine and a cosine wave

of frequency f.



The FFT

- How to keep the two correlation values for a given frequency well organized? Well, store the two values into the real and imaginary part of a complex number.
- Matlab does this very efficiently through the function fft

Periodogram

- How much of a given frequency is present in my time-series?
- SOLUTION: Take the two coefficients corresponding to a given frequency f, square them and sum them and you'll obtain the power at frequency f

The periodogram in Matlab

Let x be our time series (a Matlab array)

```
N = length(x);
X = fft(x);
S = abs(X(1:N/2)).^2;
```

• Matlab periodogram function.

Incorrect statements

- "The periodogram is the spectrum of my timeseries" (FALSE! it can be an estimate of the spectrum if the process is stationary)
- "The periodogram allows me to compute the spectrum by `pretending' that my time-series is periodical". (FALSE! If the process is periodic than it provides the true spectrum otherwise it's just an estimator)

Problems with the periodogram

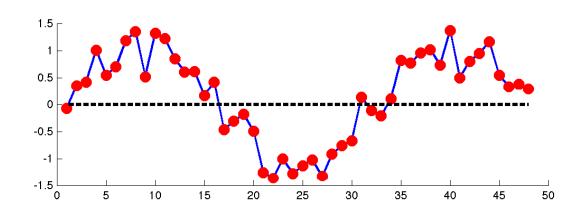
- BIAS PROBLEM: estimates done using the periodogram of spectra with high-dynamic ranges can be heavily biased.
- VARIANCE: the variance of the periodogram spectral estimates
 - is very large (as large as the value we wish to estimate)
 - does not decrease when we increase the number of points in your time analysis (i.e. you collect more data but the estimate remains as bad as it was before!).

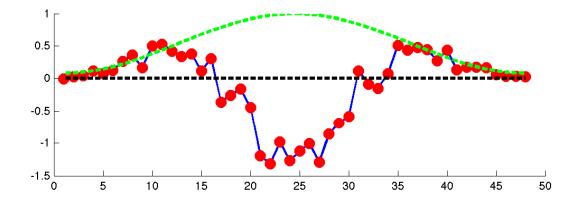


The bias problem - Tapering

- The bias in the periodogram arises because we are trying to estimate the spectrum with a very high resolution (each of the N/2 Fourier frequencies). We are stretching our data to the limit...
- To reduce the bias we need to sacrifice spectral resolution, this is done by multiplying the time series with a taper (e.g. the Hamming window).
 Unfortunately, by doing so we are throwing away some information in our data.

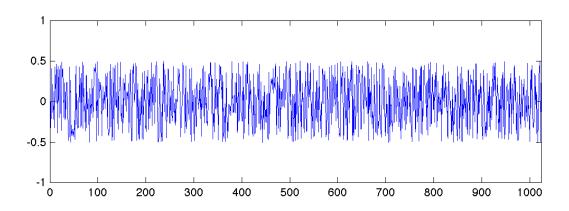
The bias problem - Tapering

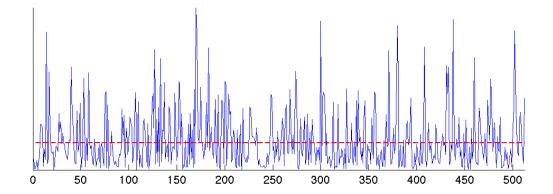






The variance problem







The variance problem

With the periodogram, when we add a data point (N -> N+1) we do not use this additional information to improve the estimates in the N/2 frequencies we were already considering. Instead, we use this new information to try estimate the spectrum at a new additional frequency. Thus adding points cannot possibly decrease the variance of our estimates.

• The only solution is to try to "average" ... but average on what?

Smoothing the periodogram [1-2]

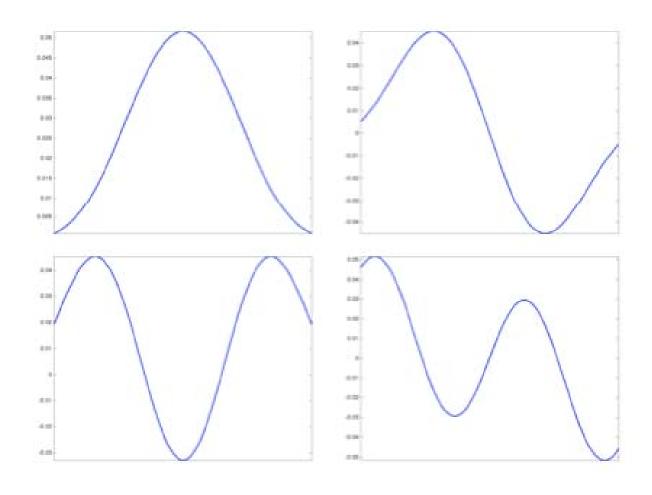
- One possibility is to smooth the spectrum in the frequency domain (lag-window estimator)
- Another possibility is to
 - cut the time-series into shorter segments,
 - compute the periodogram on the shorter (<u>tapered</u>) sequences
 - average all these estimates together.

This is what is done in the Welch's (very popular, see Matlab's **pwelch** function) and Bartlett's estimators.

Smoothing the periodogram [3]

- A third (very elegant) possibility is to try to recover some of the information lost when tapering using several different tapers each of extract a different piece of information from the data which is lost when using the other tapers.
- This is the so called multi-taper estimator
 (Matlab's pmtm function)

Multitaper





References

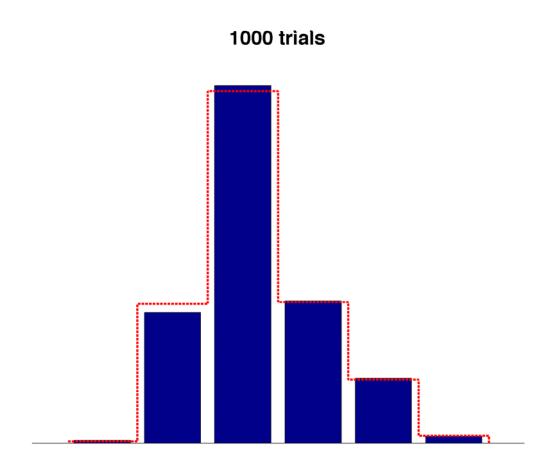
- Percivald and Walden, Spectral Analysis for Physical Applications
- Proakis, Manolakis, Digital Signal Processing

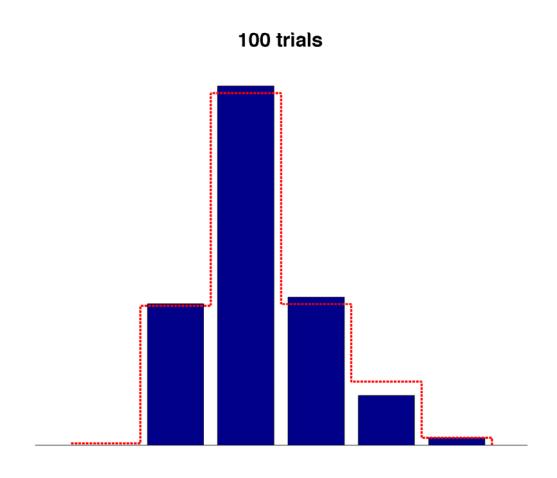
PART II – ESTIMATING MUTUAL INFORMATION

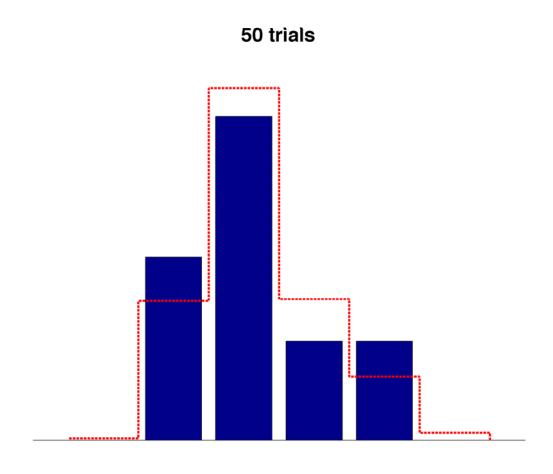
Estimating mutual-information

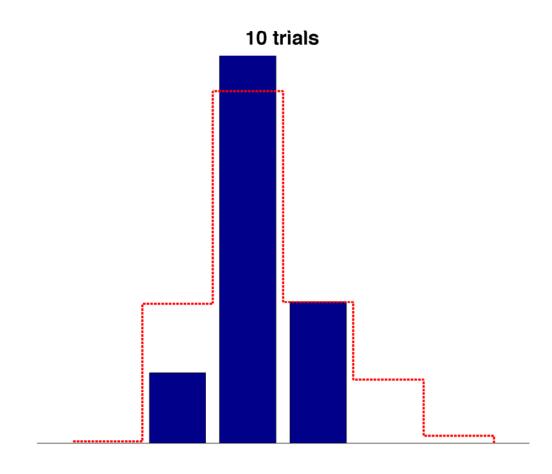
 We want to estimate the mutual information between the movie scene and spectral power

 In general to estimate mutual information we need to estimate the probability distributions.







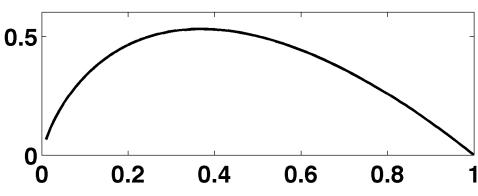


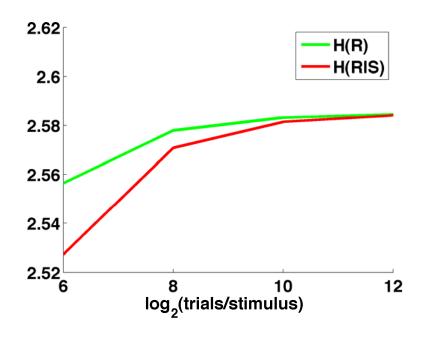
 How does this affect the estimates in mutual information? Well, mutual information is given by the difference between two

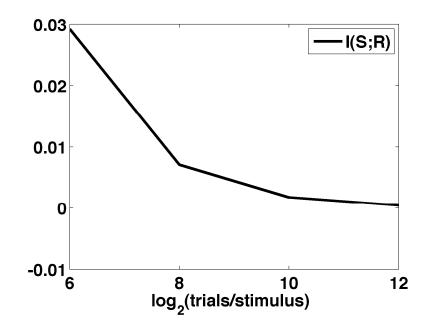
$$I(S;R) = \underset{}{\operatorname{entropies}} H(R|S)$$

and each entropylog(x) by the fixed x log(x) and each entropylog(x) by the fixed x log(x) and x log(x) and

of the form





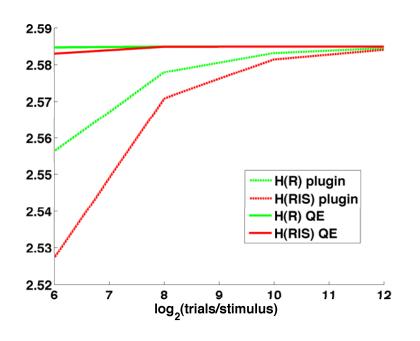


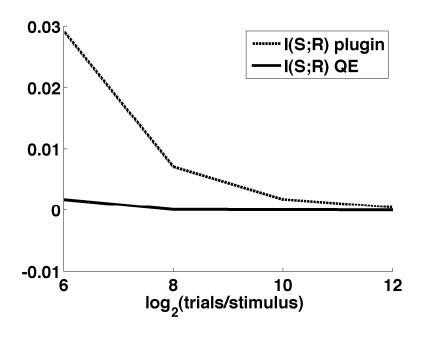
Quadratic Extrapolation

- Divide the available trials into blocks of N/2 ,
 N/4 ...
- Compute average information for N, N/2, N/4
 ... Data and fit the dependence of I on N to the above quadratic expression
- Estimate the true (N=∞) value from the bestfit expression

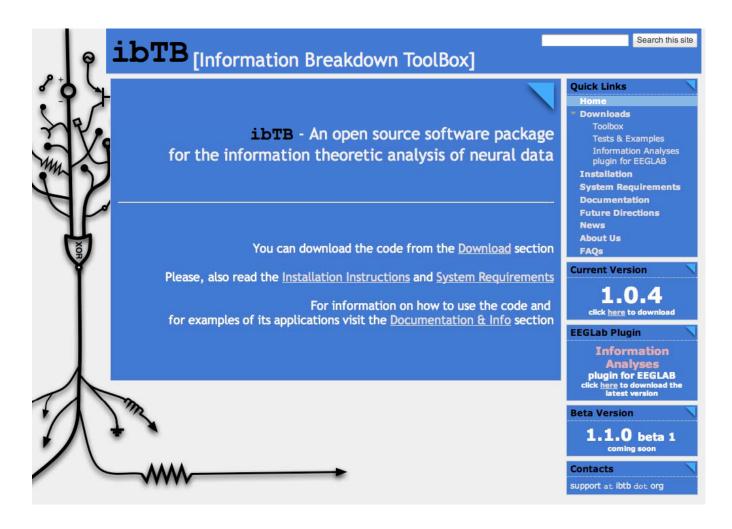
$$I_N = I_{true} + \frac{C_1}{N} + \frac{C_2}{2N^2} + \dots$$

Quadratic Extrapolation



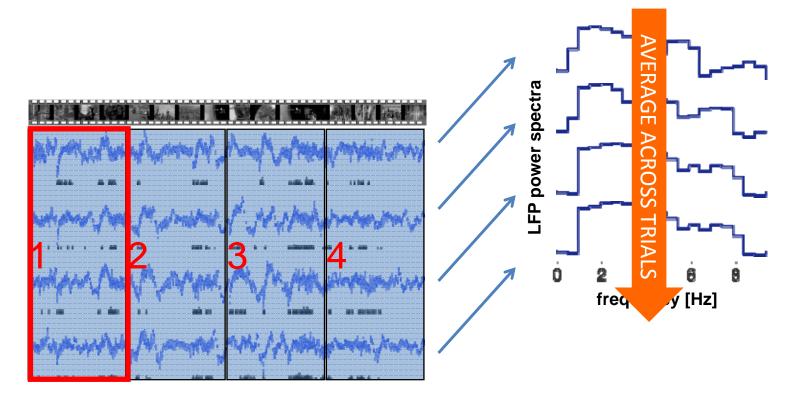


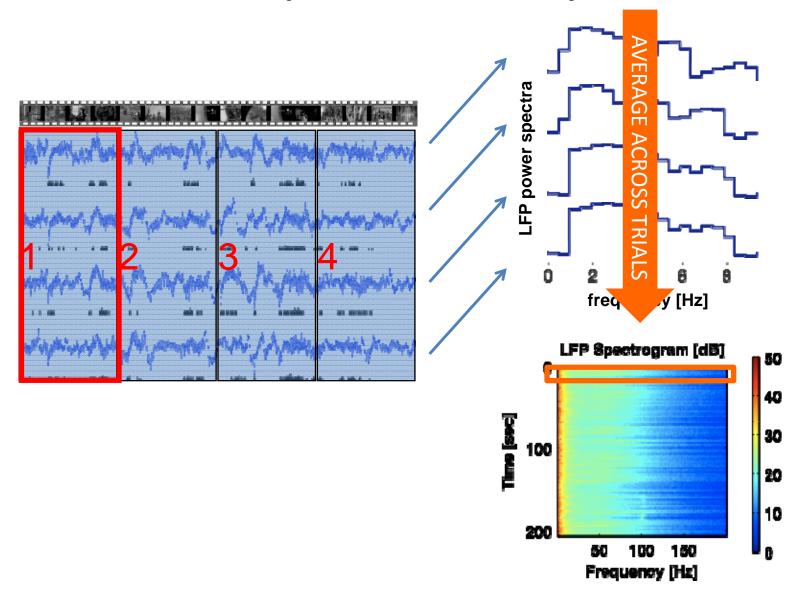
Toolbox

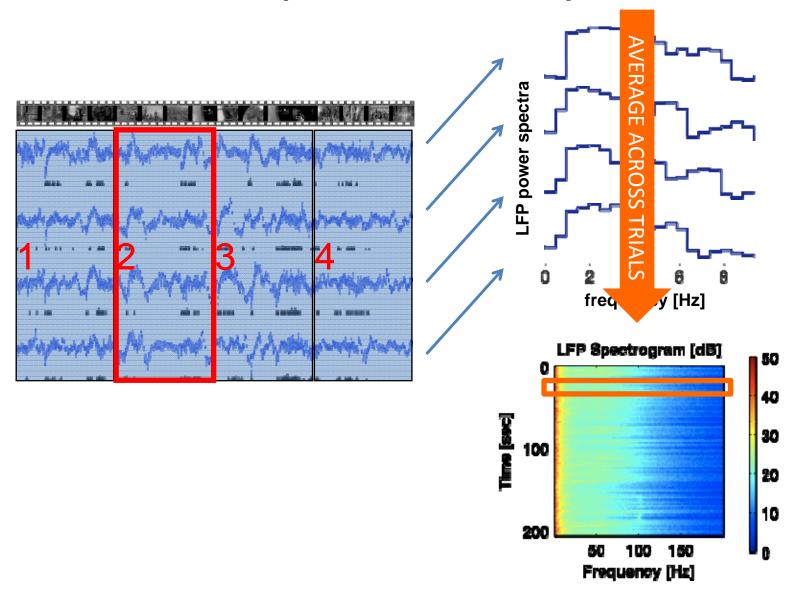


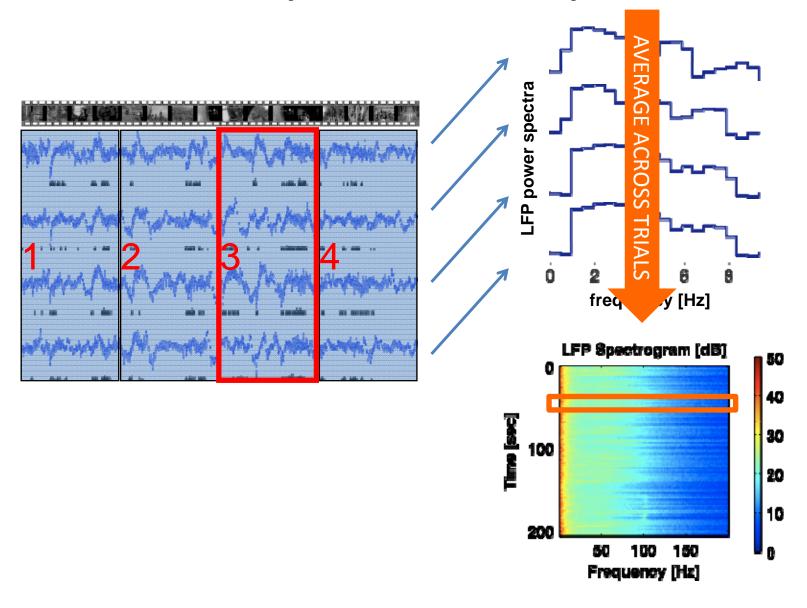
www.ibtb.org

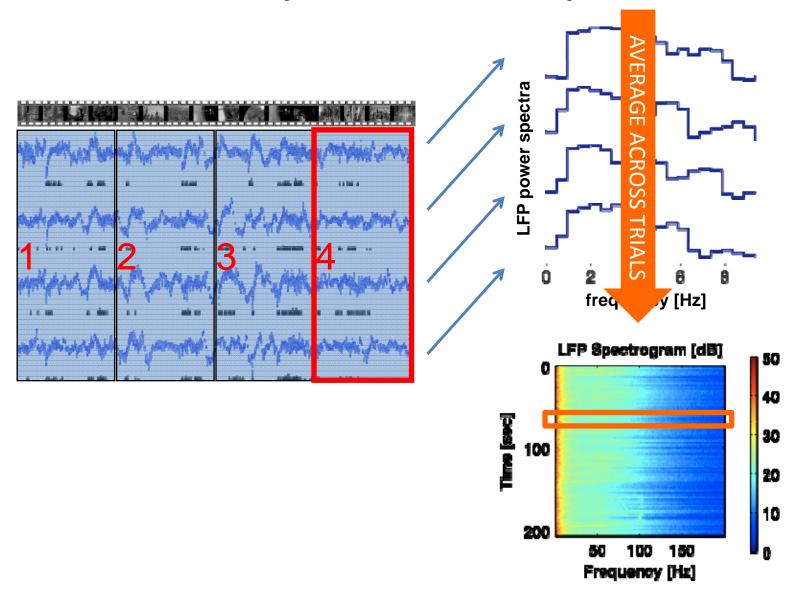
PART III – HOW TO PERFORM THE SINGLE FREQUENCY INFORMATION ANALYSIS

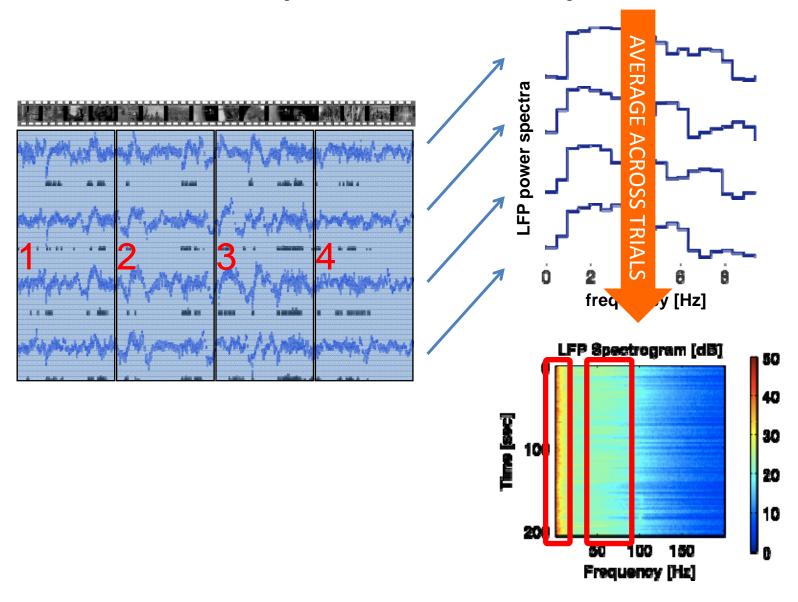




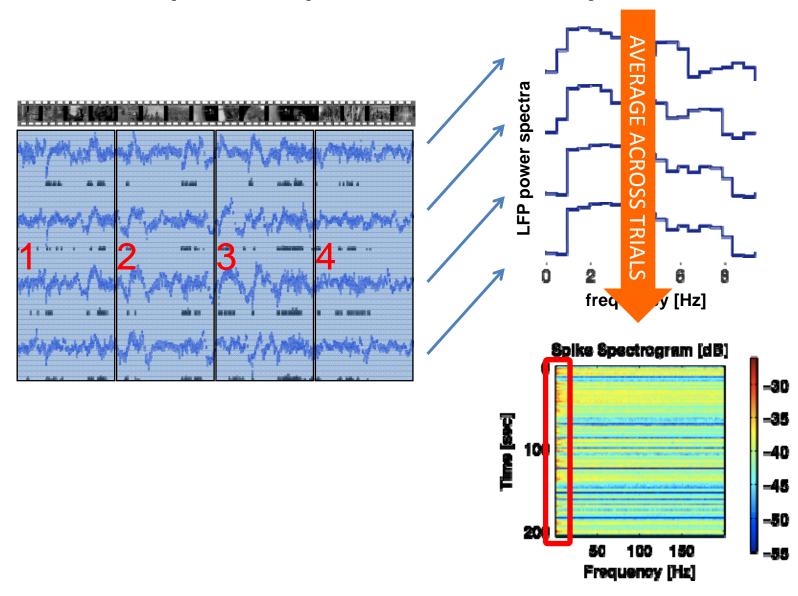






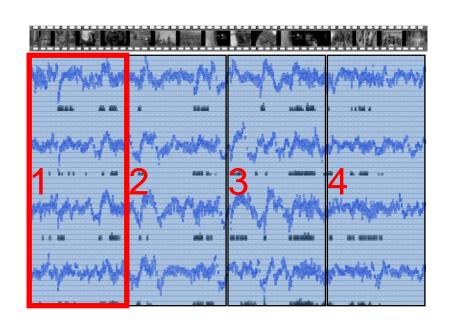


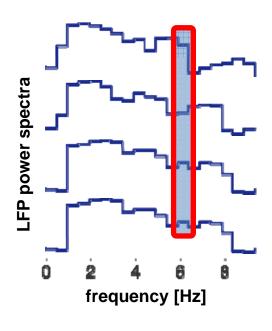
Spike spectral analysis

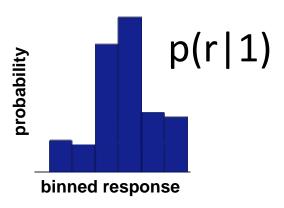


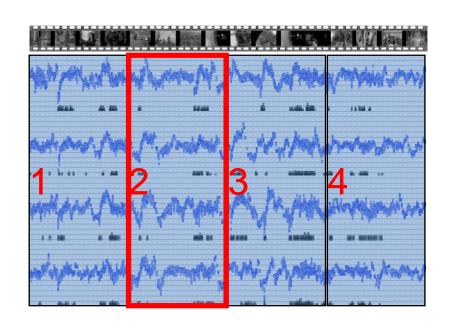
Conclusion about the spectral investigation

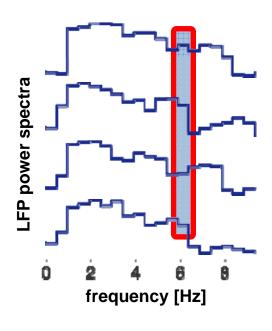
 At least for some frequencies, power really appears to convey information about which window is presented.

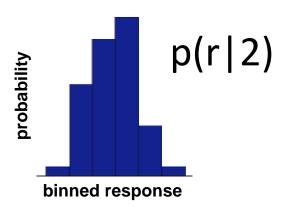


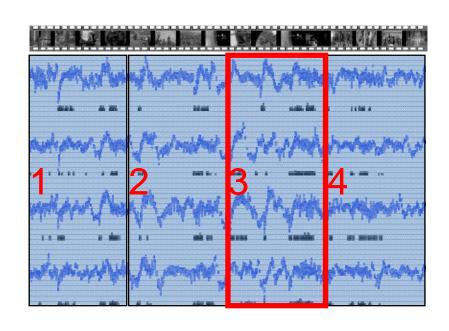


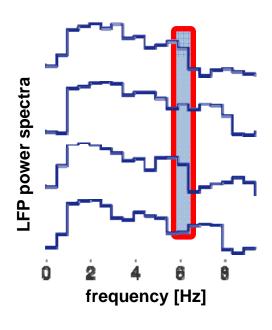


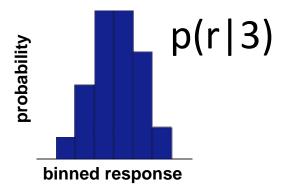


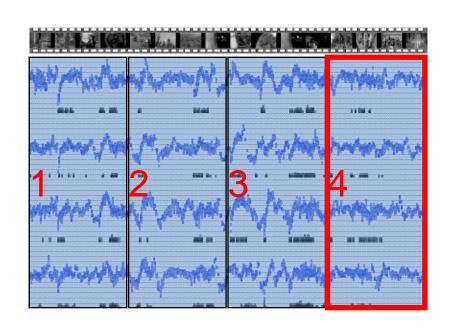


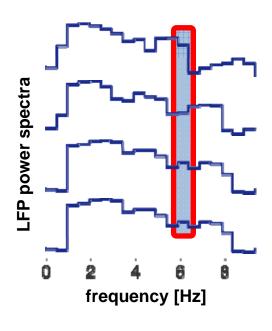


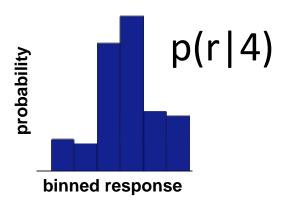








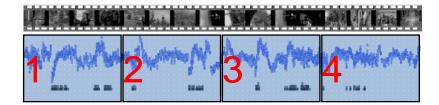




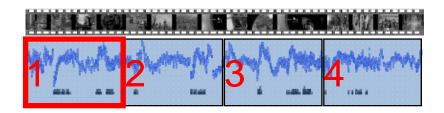
 But then we have everything we need we need to perform an information analysis for the power at each frequency!

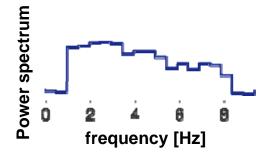
• STEP 1:

- Break each recorded sequence (trial) down into windows of chosen length (e.g. 2 seconds);
- At this point each window constitutes a "stimulus"

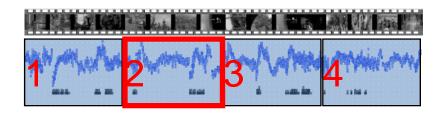


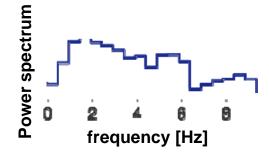
• STEP 2:



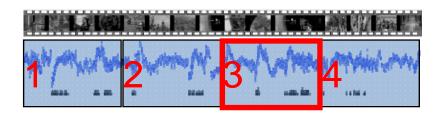


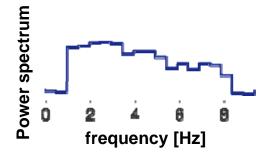
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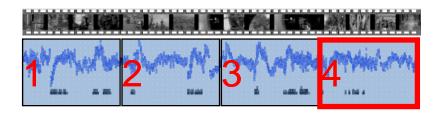


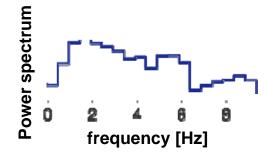
• STEP 2:





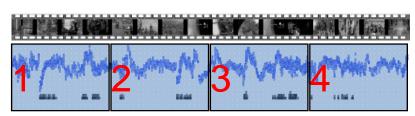
• STEP 2:

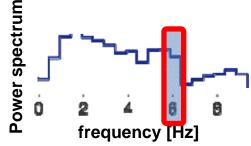




• STEP 3:

- Fix a frequency: the power values at that frequency represent the responses to the stimuli (the distinct windows)
- Feed these responses to the toolbox to compute information (use a bias correction technique).





• STEP 4:

Repeat for all available frequencies

