

Lecture 3: Signal Processing In BCIs

Introduction to Modern Brain-Computer Interface Design

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Outline

- 1. The Role of Signal Processing
- 2. Major Filter Classes
 - Spatial Filters
 - Temporal Filters
 - Spectral Filters
- 3. A Simple Neurofeedback BCI
- 4. Prediction Function Notion





3.1 The Role of Signal Processing



BCI Theory

- BCI leverages theory from a wide range of fields (Signal Processing, Machine Learning, Statistics, Neuroscience, Control Theory, Information Theory, ...)
- A given BCI may be understood from the vantage point of any of these theories
- But no single theory conveniently describes all aspects of a BCI



Signal Processing

- Digital Signal Processing is concerned with systems (a.k.a. filters) that transform one signal into another
- Linear Time-Invariant (LTI) Systems, including Spectral filters and their optimal design are one of the most developed areas
- Statistical Signal Processing and Adaptive Filtering are among the advanced areas (Kalman Filter, etc., recursive least-squares)
- Sparse Signal Processing (e.g., sparse recovery and compressive sensing) is a new branch with application to BCIs



Signal Processing

- A signal is a mapping from an index set (here discrete time) onto vectors (multichannel samples)
- From the point of view of Signal Processing, a BCI transduces the input signal x(n) (for example EEG) into a control signal y(n)
- It is defined by a *transformation rule T*

 $y(n) = \mathcal{T}[x(n)]$



Important System Types

- A system is called *static* if the value y(n) at any sample n depends only on x(n), otherwise *dynamic*.
- A system is called *causal* if the output y(n) at any time n only depends on values of x(m) for $m \le n$, otherwise *non-causal*.
- A system is called *time-invariant* if y(n) = T[x(n)]implies that y(n - k) = T[x(n - k)] for every time shift k, otherwise *time-variant*.
- A system is called *linear* if the equation $T[a_1x_1(n) + a_2x_2(n)] = a_1T[x_1(n)] + a_2T[x_2(n)]$ holds for all inputs $x_1(n)$ and $x_2(n)$ and all constants a_1 and a_2 , otherwise *nonlinear*.



BCIs Viewed as Filters

- Since BCIs are operated in real time, they are always *causal* systems
- BCIs usually perform temporal filtering, and are therefore *dynamic*
- Some BCIs are *time-invariant*, but adaptive BCIs are not
- Simple BCIs are *linear*, but the vast majority is not
- Since their output is needed at a much lower sampling rate (0.1-60Hz) than the input (250-1000Hz), they are technically *multi-rate* systems



BCI Components as Filters

- BCI components are conveniently described as filters – more so than the entire system itself
- This gives rise to several key categories of filter components







3.2 Major Filter Classes



Static Filters

• Signal Squaring: $\mathcal{T} \coloneqq y_i(n) = x_i(n)^2$

static system, useful step in calculating the variance of the signal



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Squaring



- useful later

MMMMM



Spatial Filters

- Transform a multi-channel signal X(n) such that each Y(n) depends only on X(n); most spatial filters are linear, i.e. Y(n) = MX(n) for some matrix M
- Linear spatial filters can approximately invert volume conduction and remap channel signals to approximate source signals – this is their main use in BCIs
- Examples: Re-referencing, Surface Laplacian, Independent Component Analysis (ICA), Common Spatial Patterns (CSP) – more later



Spatial Filters Visualized

 Spatial filters designed to *recover motorcortex source activity*, calculated via the Common Spatial Patterns algorithm





Spatial Filters vs. Forward Projections

 Spatial filters are *not* the same as forward projection maps of some source signal – they are the inverse operation





Temporal Filters

- Transform a multi-channel signal X(n) such that each channel $y_i(n)$ in Y(n) depends only on the channel $x_i(n)$
- They are conceptually orthogonal to spatial filters
- Examples include time windowing, wavelet transform, etc.
- Special case: Spectral filters



Example Temporal Filters

• Moving Average:

$$\mathcal{T} := y_i(n) = \frac{1}{m} \sum_{k=0}^{m-1} x_i (n-k)$$

- Effectively a smoothing (low-pass) operator
- In fact a simple example of a spectral filter





Spectral Filters

- Temporal filters that are designed for their effects on the *spectrum* of the signal
- **Spectrum of a signal**: a representation of the signal as a sum of *N* sinusoidal components,

 ΛT

$$s(n) = \sum_{k=1}^{N} A_k \sin(\omega_k nT + \phi_k)$$

where A_k is the amplitude of each sinusoid and ϕ_k is its phase.



Spectral Filters

• An equivalent (more common) representation is the Fourier Series representation

$$s(n) = \sum_{k=0}^{N-1} A_k e^{j2\pi kn/N}$$

where A_k is now complex-valued and represents both the amplitude and the phase

• This relies on the Euler formula

$$e^{\pm j\phi} = \cos\phi \pm j\sin\phi$$



Spectral Filters

- Examples include: High-pass, Low-pass, Bandpass filters, Notch filters
- Their main utility in BCIs is to isolate oscillations or ERPs of interest

Bandpass Filter (2-30Hz)





A Key Spectral Filter

• FIR (Finite Impulse Response) Filter:

$$\mathcal{T} \coloneqq y_i(n) = \sum_{k=0}^m b_k x_i (n-t)$$

- Performs a *convolution* between signal and kernel
- The trick lies in the coefficients ("kernel") b_k
- Can implement any linear time-invariant spectral filter
- Moving average is a special case



Filter Implementations







Filter Implementations





Filter Implementations





Minimum-Phase Filter Design

• Linear-Phase Lowpass (uniform lag):



Minimum-Phase Lowpass (minimal lag):





FIR Filter Design in MATLAB

- Various criteria for filter kernel design (given a desired frequency response):
 - Least-Squares error: firls
 - Minimax error (Parks-McClellan): firpm
 - Using the Fourier transform: fir2
- Choice of a reasonable filter order (length):
 firpmord
- Minimum-phase filter design (using Cepstral analysis): rceps



Other Filters

- Spatio-Temporal Filters are also used, but have too many degrees of freedom to be handdesigned; usually the result of an adaptive procedure
- **Spectral Transforms**, which transform between the time and spectral representation of a signal are used frequently as intermediate stages
- Rate-changing Filters such as resampling are useful to manage computational costs





3.3 A Simple Neurofeedback BCI



A Simple Neurofeedback BCI

- Feed back the amplitude of a brain idle oscillation (e.g. 10 Hz alpha for relaxation) to the user/subject
- Also other processes conceivable



source: psychful.com



A Simple Neurofeedback BCI

- Feed back the amplitude of a brain idle oscillation (e.g. 10 Hz alpha for relaxation) to the user/subject
- Can be implemented using discussed tools:







3.4 Prediction Function Notion



Alternative to the Signal Processing Framework

• A BCI with a limited memory of the past could also be viewed as a mathematical mapping *f*:





Alternative to the Signal Processing Framework

• A BCI with a limited memory of the past could also be viewed as a mathematical mapping *f*:

y= "subj. excited" (+1)
 "subj. not excited" (-1)

- The functional form is arbitrary, for example
 y = sign(var(WX) + b)
- The mapping involves unknown parameters, here W and b



Functional Form

- Reflects the relationship between observation (data segment X) and desired output (cognitive state parameter y)
- Based on some assumed generative mechanism (forward model) – or ad hoc



• Note: Functional form is the inverse mapping!



Core Ingredient: Spatial Filter

- Linear inverse of volume conduction effect
 - X = AS (forward)
 - S = WX (inverse)





Full Examples (Details Later)

• Inverse mapping from filtered source time courses to latent cognitive state, e.g.:

$$y = \theta \operatorname{vec}(WX) + b$$
 (linear)

$$y = \theta \operatorname{vec}(|(WX)T|) + b$$
 (nonlinear...)



Neurofeedback BCI in Functional Style

 Performed as a mapping of a sliding window X onto the output y:

$$f(\mathbf{X}) := y = \log var(\mathbf{XT})$$

 T implements a temporal filter, written as a matrix multiplication (each column = shifted filter kernel)



In Comparison...

- Main drawback of the pure mathematical form compared to the signal processing approach:
 - The entire input window X is *re-processed* for each desired output value y
 - Especially bad if the window X moves only by a few samples between evaluations of f
 - In contrast, most signal processing methods are incremental or recursive (e.g. FIR or IIR)
- Main benefit is the relative conceptual simplicity



In Combination

- Both frameworks are complementary, rather than contradictory, and are in practice often used *in combination*
- Prediction function is queried on demand





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Neurofeedback Using the Combined Approach

- Computationally costly spectral filtering is done in the signal processing portion
- Lightweight predictive mapping is done at lower rate in the functional portion





Outside View

 Implemented this way, BCIs act as an *oracle* that consumes one or more multi-channel signals and can respond to queries about a pre-defined question





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- Implemented this way, BCIs act as an *oracle* that consumes one or more multi-channel signals and can respond to queries about a pre-defined question
- Note: in modern BCIs the output is often a discrete probability distribution







L3 Questions?