T-61-181 Biomedical Signal Processing

Sections 4.5.3 – 4.6.2

Learning basis vectors and weights for EEG signals

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Finding optimal basis function to denoise

- 1. Karhunen-Loeve Expansion
- 2. Interprtaion as Linear, Time-Variant Filtering
- 3. Damped Sinusoids : Finding Optimal function

Adaptive Analysis Using Basis Functions

- 1. Instantaneous LMS algorithm
- 2. Block LMS algorithm

Karhunen-Loeve Expansion – optimal basis function

To design the basis function φ_k such that the signal part is efficiently represented with small number of functions

Derivation:

Decompostion of signal x in to two sums representing signal and noise

$$\mathbf{x} = \sum_{k=1}^{K} w_k \varphi_k + \sum_{k=K+1}^{N} w_k \varphi_k = \hat{\mathbf{s}} + \hat{\mathbf{v}}.$$

Aim is to find set of φ_k such that resembles original signal s as closely as possible.

Can be done by minimising the noise power estimate:

$$\mathcal{E} = E\left[\hat{\mathbf{v}}^T \hat{\mathbf{v}}\right] = E\left[(\mathbf{x} - \hat{\mathbf{s}})^T (\mathbf{x} - \hat{\mathbf{s}})\right]$$

PROOF

considering that the actual signal is composed of x = s + v the noise power estimate becomes.

$$\mathcal{E} = E\left[(\mathbf{s} - \hat{\mathbf{s}})^T (\mathbf{s} - \hat{\mathbf{s}})\right] + 2E\left[(\mathbf{s} - \hat{\mathbf{s}})^T \mathbf{v}\right] + E\left[\mathbf{v}^T \mathbf{v}\right]$$

$$E[\mathbf{v}\mathbf{v}^T] = \mathbf{R}_v = \sigma_v^2 \mathbf{I}$$

as noise is zero-mean with correlation matrix $\sigma_v^2 \mathbf{I}$

$$E\left[(\mathbf{s} - \hat{\mathbf{s}})^T \mathbf{v}\right] = 0 - E\left[\left(\sum_{k=1}^K w_k \varphi_k\right)^T \mathbf{v}\right]$$
$$= -E\left[\left(\sum_{k=1}^K (\varphi_k^T \mathbf{x}) \varphi_k\right)^T \mathbf{v}\right]$$
$$= -E\left[\left(\sum_{k=1}^K \varphi_k^T (\mathbf{s} + \mathbf{v}) \varphi_k\right)^T \mathbf{v}\right]$$

as signal noise are assumed to be uncorrelated.

$$= -\sum_{k=1}^{K} \operatorname{tr} \left(E\left[\mathbf{v}^{T} \boldsymbol{\varphi}_{k} \boldsymbol{\varphi}_{k}^{T} \mathbf{v} \right] \right)$$
$$= -\frac{K}{N} E\left[\mathbf{v}^{T} \mathbf{v} \right]$$
$$= -K \sigma_{r}^{2}.$$

$$\mathcal{E} = E\left[\hat{\mathbf{v}}^T \hat{\mathbf{v}}\right] = E\left[(\mathbf{s} - \hat{\mathbf{s}})^T (\mathbf{s} - \hat{\mathbf{s}})\right] + (N - 2K)\sigma_v^2.$$

Optimal basis function and correlation matrix of data

We start frim the equation of noise power estimate again:

$$\mathcal{E} = E\left[\hat{\mathbf{v}}^T \hat{\mathbf{v}}\right] = E\left[\left(\sum_{k=K+1}^N w_k \varphi_k\right)^T \left(\sum_{l=K+1}^N w_l \varphi_l\right)\right] = \sum_{k=K+1}^N E\left[w_k^2\right]$$
$$= \sum_{k=K+1}^N \varphi_k^T \mathbf{R}_x \varphi_k \qquad \text{since } w_k^2 = \varphi_k^T \mathbf{x} \mathbf{x}^T \varphi_k = \varphi_k^T \mathbf{R}_x \varphi_k$$

Minimisation by the use of Lagrange multiplier to ensure that orthonoramilty of φ is mentained. we get the function to be minimised as :

$$\mathcal{L} = \sum_{k=K+1}^{N} \varphi_k^T \mathbf{R}_x \varphi_k + \sum_{k=K+1}^{N} \lambda_k (1 - \varphi_k^T \varphi_k)$$

lagrange multipliers

MSE as Sum of Eigen values

The minimization of the above function with lagrange multipliers can eb done by taking gradient with respect to φ_{k} and setting it to zero, which gives:

$$\nabla_{\varphi_k} \mathcal{L} = \mathbf{R}_x \varphi_k - \lambda_k \varphi_k = 0, \quad \text{or} \quad \mathbf{R}_x \varphi_k = \lambda_k \varphi_k, \quad k = K + 1, \dots, N$$

so mean square error can be represented in terms of lagrange coefficients or eigen values λ_k

$$\mathcal{E} = \sum_{k=K+1}^{N} \varphi_k^T \mathbf{R}_x \varphi_k$$
$$= \sum_{k=K+1}^{N} \varphi_k^T (\lambda_k \varphi_k) = \sum_{k=K+1}^{N} \lambda_k$$

So is minmised when *N- K* smallest eigenvalues are chosen

or

When signal s is represented in terms of basis vector corresponding to largest K eiegen vales

Interpretations



It can be viewed as finding direction of maximum variance in n dimensional space

Interpretation as Linear, Time Variant Filtering

when signal is estimated as means of different basis function i.e.

$$\dot{\mathbf{s}}_{i} = \boldsymbol{\Phi}_{s} \boldsymbol{\Phi}_{s}^{T} \mathbf{x}_{i} = \sum_{k=1}^{K} \sum_{l=0}^{N-1} \varphi_{k}(n) \varphi_{k}(l) \mathbf{x}_{i}(l)$$

$$= \sum_{l=0}^{N-1} g(l, n) \mathbf{x}_{i}(l), \quad n = 0, \dots, N-1$$

where

$$g(l,n) = \sum_{k=1}^{K} \varphi_k(l) \varphi_k(n), \quad l,n = 0, \dots, N-1$$

Interpretation in frequency domain



functions

Modeling with Damped Sinusoids

Sinusoidal basis function with exponential damping can be used to make a variety of basis functions, such as data x can be represented as :



Since x(n) is generally real valued signal, so data can be represented by damped cosines because sum of conjugate complex pairs is cosine.

$$x(n) = \sum_{k=1}^{K/2} 2w_k e^{\rho_k n} \cos(\omega_k n + \phi_k)$$

Pros and cons of damped sinusoid as basis function





.Damping add an extra degree of freedom , so fewer basis function represent the data

No orthonormality

Non linearity of equations makes it tough to find the solutions for best basis function

Prony Method

The model
$$x(n) = \sum_{k=1}^{K} w_k e^{\rho_k n} e^{j(\omega_k n + \phi_k)}$$

Can be represented as $x(n) = \sum_{k=1}^{K} h_k z_k^n$. Where $h_k = w_k e^{j\phi_k}$
 $z_k = e^{\rho_k + j\omega_k}$

This model can be viewed as homogenous soultion to a linear difference equation with fixed parameters. **i.e.**

$$x(n) + a_1 x(n-1) + \ldots + a_k x(n-k) = 0$$

In prony's original method it assumed that number of available samples is equal to unknown parameters, so the difference equation is valid for n = k, ... 2k-1

The difference equation can be represented as k x k matrix equation as:

$$\begin{bmatrix} x(K-1) & x(K-2) & \cdots & x(0) \\ x(K) & x(K-1) & \cdots & x(1) \\ \vdots & \vdots & \ddots & \vdots \\ x(2K-2) & x(2K-3) & \cdots & x(K-1) \end{bmatrix} \begin{bmatrix} a_1 \\ a_2 \\ \vdots \\ a_K \end{bmatrix} = - \begin{bmatrix} x(K) \\ x(K+1) \\ \vdots \\ x(2K-1) \end{bmatrix}$$

On solving the above equation we get values of $a_1 \dots a_k$ which can be used to find the roots of the polynomial

$$A(z) = \sum_{l=0}^{K} a_l z^{K-l} = \prod_{k=1}^{K} (z - z_k)$$

The roots can be used to determine the parameters $h_1, \ldots h_k$ from the eq.

$$\begin{bmatrix} z_1^0 & z_2^0 & \cdots & z_K^0 \\ z_1^1 & z_2^1 & \cdots & z_K^1 \\ \vdots & \vdots & \ddots & \vdots \\ z_1^{K-1} & z_2^{K-1} & \cdots & z_K^{K-1} \end{bmatrix} \begin{bmatrix} h_1 \\ h_2 \\ \vdots \\ h_K \end{bmatrix} = \begin{bmatrix} x(0) \\ x(1) \\ \vdots \\ x(K-1) \end{bmatrix}$$

Prony method for real problems: least square Prony method

When we want to use less parameters k than available number of observations N we have relax the requirement of difference equation as given below

$$x(n) + a_1 x(n-1) + ... + a_k x(n-k) = e(n)$$
 - error

And to minimise the error e(n) w.r.t the parameters $a_1, \ldots a_k$

The parameters h1 . . .hk is determined by minimizing the function:

$$\|\mathbf{x}_N - \mathbf{Z}_N \mathbf{h}_K\|_2^2 = (\mathbf{x}_N - \mathbf{Z}_N \mathbf{h}_K)^T (\mathbf{x}_N - \mathbf{Z}_N \mathbf{h}_K).$$

And the least square solution comes out to be

 $\mathbf{h}_K = (\mathbf{Z}_N^H \mathbf{Z}_N)^{-1} \mathbf{Z}_N^H \mathbf{x}_N.$

Adaptive analysis using Basis Functions

Instanteneous LMS Algorithm

The observed signal is made from concatenation of successive Evoked potentials $\mathbf{x}_i = [x((i-1)N), \dots, x(iN-1)]^T$

$$x(n) = x_{\lfloor \frac{n}{N} \rfloor + 1} \left(n - \lfloor \frac{n}{N} \rfloor N \right)$$

The siganl estimate is calcukated from truncated series expansion but this time with time varying weight vector w(n). For this purpose minimum of MSE is done at every sample. For this the differenital of the function

$$\mathcal{E}_{\mathbf{w}}(n) = E\left[(x(n) - \boldsymbol{\varphi}_s^T(n)\mathbf{w}(n))^2\right]$$

is taken to yeild : - $\nabla_{\mathbf{w}} \mathcal{E}_{\mathbf{w}}(n) = -2E[(\mathbf{x}(n) - \boldsymbol{\varphi}_{s}^{T}(n)\mathbf{w}(n)) \boldsymbol{\varphi}_{s}(n)]$

using the gradient to update weights and using mean μ instead of expectation the update formula is :

$$\mathbf{w}(n+1) = \mathbf{w}(n) + \mu(x(n) - \boldsymbol{\varphi}_s^T(n)\mathbf{w}(n))\boldsymbol{\varphi}_s(n)$$
$$= (\mathbf{I} - \mu\boldsymbol{\varphi}_s(n)\boldsymbol{\varphi}_s^T(n))\mathbf{w}(n) + \mu x(n)\boldsymbol{\varphi}_s(n)$$

which when expressed in terms of initial weight vector w(0) can be written as

$$\mathbf{w}(n) = \mathbf{F}_0(n-1)\mathbf{w}(0) + \mu \sum_{m=0}^{n-1} x(m)\mathbf{F}_{m+1}(n-1)\varphi_s(m)$$

where

$$\mathbf{F}_m(n) = \begin{cases} \prod_{j=m}^n \left(\mathbf{I} - \mu \varphi_s(j) \varphi_s^T(j) \right) & n \ge m \\ \mathbf{I} & m < n. \end{cases}$$



A representation of adaptive analysis using LMS algorithm

The Block LMS algorithm

Its a extension of singla trial LMS method using truncated series expansion. the error function is quiet simmilar

$$\mathcal{E}_{\mathbf{w}_{i-1}} = E\left[(\mathbf{x}_i - \Phi_s \mathbf{w}_{i-1})^T (\mathbf{x}_i - \Phi_s \mathbf{w}_{i-1}) \right]$$

but w_{i-1} is used instead of w_i

so the updateing of weights is done by

$$w_{i} = (1-\mu)w_{i-1} + \mu F_{s}^{T} x_{i} \quad or \quad w_{i} = w_{i-1} + \mu F_{s}^{T} (x_{i} - F_{s} w_{i-1})$$