Local Polynomial Modelling of Time-varying Autoregressive Processes and its Application to the Analysis of Event-related Electroencephalogram

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Abstract—This paper proposes a new method for identification of time-varying autoregressive (TVAR) models based on local polynomial modeling (LPM) and applies it to investigate the dynamic spectral information of event-related electroencephalogram (EEG). The proposed method models the TVAR coefficients locally by polynomials and estimates those using least-squares estimation with a kernel having a certain bandwidth. A data-driven variable bandwidth selection method is developed to obtain the optimal bandwidth, which minimizes the mean squared error (MSE). Simulation results show that the LPM-based TVAR identification method outperforms conventional methods for different scenarios. The advantages of the LPM method make it a useful high-resolution time-frequency analysis (TFA) technique for nonstationary biomedical signals like EEG. Experimental results show that the LPM method can reveal more meaningful time-frequency characteristics than wavelet transform.

I. INTRODUCTION

Time-varying autoregressive (TVAR) models are widely used to describe the dynamics of nonstationary signals, such as speech signals, communication signals, and biomedical signals [1], [2]. A nonstationary signal \( s(t) \) can be expressed in the form of a TVAR model as follows:

\[
s(t) = \sum_{l=1}^{L} a_l(t)(t - kT) + \varepsilon(t),
\]

where \( a(t) = [a_1(t), a_2(t), \ldots, a_L(t)]^T \) is the TVAR coefficient vector, \( L \) is the order of the model, \( T_s \) is the sampling period, and \( \varepsilon(t) \) is a zero mean additive white Gaussian process with variance \( \sigma^2(t) \). Accurate identification of TVAR models, i.e., estimation of the TVAR coefficients \( a(t) \), is crucial to understand the dynamics of signals and to predict future observations. Generally, two classes of TVAR identification methods are common in practice [2], and they are 1) adaptive filtering and Kalman filtering (KF); 2) basis expansion modelling (BEM). Although these methods usually offer efficient implementation and different tradeoffs between performance and complexity, their performances are often dependent on prior information and model parameters [1], [2]. Therefore, the estimation accuracy of these methods is sometimes limited when prior knowledge is vague or parameters are incorrectly specified [2].

In this paper we propose a new TVAR identification method based on a local polynomial modeling (LPM) approach, which does not require explicit information of the coefficient variations. It can achieve a good tracking performance and a better flexibility in a variety of experimental conditions. The proposed LPM method models the TVAR coefficients locally by a set of polynomials and a window or kernel having a certain bandwidth. The estimation of TVAR coefficients is then reduced to the estimation of polynomial coefficients, which can be performed using the least-squares (LS) technique [3]-[7]. Since the TVAR coefficients may vary considerably over time, it is crucial to choose a proper local bandwidth to achieve the best bias-variance tradeoff. For slow-varying coefficients, we would like the bandwidth to be large to reduce the estimation error. For fast-varying coefficients, a small bandwidth is desirable in order to reduce the bias error. A data-driven bandwidth selection technique for LPM is therefore developed, where the bandwidths are determined locally and adaptively by minimizing the approximated mean squared error (MSE), which is the sum of squared bias and variance. Hence, the proposed LPM-based TVAR identification method can obtain good tracking performances for both slowly-varying and fast-varying coefficients by employing variable bandwidths.

Next, we propose to apply the LPM-based TVAR identification method for time-frequency analysis (TFA) of event-related electroencephalogram (EEG). Event-related EEG is the stimulus-induced electrical activities measured on the scalp, and it turns out to be a non-stationary process, because the states of brain regions change rapidly on a time scale of milliseconds [8]. Previous studies showed that event-related EEG consists of a large number of components, such as event-related potentials (ERPs), event-related desynchronization (ERS/ERD), and other oscillations and potentials [8], [9]. TFA techniques have been widely adopted in EEG study, because the time-varying spectral components can effectively represent ongoing EEG rhythmic activities, providing useful information for discovering the nature and timing of sensory and cognitive processes.

Currently the most popular TFA method for EEG is the wavelet transform (WT) [10]. The superiority of WT is that it addresses the time-frequency resolution tradeoff problem by applying a short window at high frequency and a long window at low frequency. However, WT has a degraded frequency resolution for high-frequency components and a degraded time resolution for low-frequency components. Therefore, WT may not able to reveal useful components in the whole time-frequency plane and cannot acquire accurate temporal information on rapid (maybe event-related) changes of EEG. Another type of TFA methods (parametric methods) describe an EEG signal by a parametric model, and the time-varying power spectral density (PSD) of the signal can be obtained from the estimated model coefficients, hence providing a time-frequency distribution (TFD) of the EEG. Parametric methods generally have a high frequency
resolution than nonparametric methods like WT and Lomb periodogram [11]-[13]. In EEG study, TVAR models and KF are respectively the most popular parametric model and identification method. However, KF is sensitive to model parameters so that it often leads to a long tracking lag or a large estimation variance (time-frequency resolution tradeoff) [11]. The proposed LPM-based TVAR identification method is well-suited for TFA of EEG because its bandwidth (time-scale) is adaptively selected so that the TFD can achieve high time-resolution without any loss of frequency resolution. EEG signals recorded in a visual oddball paradigm are tested and the experimental results show that the LPM method can reveal more meaningful time-frequency components than WT.

The rest of the paper is organized as follows. In Section II, the LPM for TVAR models is introduced. The adaptive bandwidth selection method for the LPM-based TVAR identification is developed in Section III. The proposed method is further extended to TFA of EEG in Section IV. Experimental results are presented in Section V. Finally, conclusions are drawn in Section VI.

II. LPM FOR IDENTIFICATION OF TVAR MODEL

In the TVAR model of (1), the \( k \)-th coefficient \( a_k(t) \) can be modelled locally at \( t=t_0 \) as a \( p \)-th order polynomial [3]-[6]:

\[
a_k(t) = \sum_{j=0}^{p} \alpha_{kj}(t_0)(t-t_0)^j,
\]

where \( \alpha_{kj}(t_0) \) are the polynomial coefficients. These polynomial coefficients can be estimated locally by maximum likelihood (ML) estimation. Since the additive noise is zero mean and white Gaussian distributed, the ML estimation is equivalent to minimizing a locally weighted LS criterion as follows:

\[
\min_{\beta} \sum_{t_0} \left[ s(t) - \sum_{j=0}^{p} \beta_{kj}(t_0)(t-t_0)^j \right]^2 K_0(t-t_0),
\]

where \( s(t) \), \( \beta_{kj}(t_0) \), \( K_0(t-t_0) \) is a kernel which controls the number of neighboring samples around \( t_0 \) used to estimate \( \beta_{kj}(t_0) \). The kernel \( K_0(t-t_0) \) is obtained by scaling a basis kernel \( K_0(t-t_0) \) using a bandwidth of \( h_0 \).

Next, we rewrite (3) more compactly in a matrix form as

\[
\min_{\beta} \left( y - X \beta \right)^T W \left( y - X \beta \right),
\]

where \( y = [s_1, s_2, \ldots, s_n]^T \), \( W = \text{diag}(K_0(t_0-t_0)) \),

\[
X = \begin{bmatrix} X^T(t_1) & \ldots & X^T(t_n) \\ \vdots & \ddots & \vdots \\ X^T(t_n) & \ldots & X^T(t_1) \end{bmatrix}
\]

with \( X^T(t) = [x(t), x(t-t_0), x(t-2t_0), \ldots] \) and \( s(t) = [s(t-T), \ldots, s(t-LT)]^T \), and \( \beta = [\beta_{00}(t_0), \beta_{10}(t_0), \ldots] \).

The LS solution to (4) is

\[
\hat{\beta}(t_0) = (X^T W X)^{-1} X^T W y,
\]

and finally the \( k \)-th TVAR coefficient at time instant \( t_0 \) is obtained as \( \hat{a}_k(t_0) = \hat{\alpha}_{kj}(t_0) = \hat{\beta}_{kj}(t_0) \).

The TVAR model can be viewed as a special case of a time-varying linear regression model, where \( s(t) \), \( s(t-kT) \), and \( \alpha(t) \) are respectively the observations, explanatory variables and regression coefficients. In [7], LPM has been proposed for estimating the regression parameters of the time-varying linear regression model, and the asymptotic properties of bias and variance have been derived. In the content of TVAR model, the asymptotic bias and variance of \( \hat{\beta}_{kj}(t_0) \) are respectively expressed as:

\[
\text{Bias}(\hat{\beta}_{kj}(t_0)) = \frac{\hat{S}_{kj}(0)p!}{n^{p+1}} s(t_0-kT),
\]

\[
\text{Var}(\hat{\beta}_{kj}(t_0)) = \frac{\hat{S}_{kj}(0)p!}{nh^{p+1}} s(t_0-kT),
\]

where \( \hat{S}_{kj}(0) \) and \( \hat{S}_{kj}(0) \) are two quantities determined by the basis kernel \( K_0(t-t_0) \) and indices \( j \) and \( k \), and \( \hat{\beta}_{kj}(t_0) \) is the \( (p+1) \)-th derivative of \( \hat{S}_{kj}(0) \). The derivation and details of (6) and (7) can be found in [7]. It can be seen from (6) and (7) that, when \( h \) increases, the bias will increase while the variance will decrease. Hence, there exists a locally optimal bandwidth \( h_0^* \) for estimating \( \hat{\beta}_{kj}(t_0) \), and \( h_0^*(k, t_0) \) should minimize the MSE. As some of the quantities in (6) and (7) are difficult to be calculated directly, the optimal bandwidth is difficult to be estimated accurately. We next introduce an empirical method to select the optimal bandwidth from a set of possible bandwidths.

III. ADAPTIVE BANDWIDTH SELECTION FOR LPM

In [7], we have developed an effective bandwidth selection method for time-varying linear regression models. This method is also applicable to the TVAR models with slight modifications. Given a finite set of bandwidth parameters in a geometric grid, say,

\[
H = \{h_j \mid h_j = h_0 h_j^*, \ j = 0, \ldots, J-1\},
\]

where \( h_0 > 1 \) is a step factor, \( h_0 > 0 \) is a base bandwidth, and \( J \) is the number of bandwidths, the empirical bandwidth selection method in [7] approximates the bias, variance, and MSE values of each bandwidth and determines an optimal bandwidth as the one that minimizes the approximated MSE.

Firstly, the bias of \( \hat{\beta}(t_0) \) can be estimated using a Taylor’s expansion with an order \( p+\rho \text{ex} \):

\[
\hat{b}_j(t_0) = (X^T W X)^{-1} X^T W r,
\]

where \( r \) is an \( n \times 1 \) vector with \( \sum_{i=0}^{p+\rho_j} \beta_{kj}(t_0)(t-t_0)^i s(t_k T_i) \) as its \( j \)-th element. \( \hat{b}_j(t_0) \) can be estimated by fitting a polynomial of degree \( p+\rho_j \), where \( p+\rho_j \) is an excess order [3]. However, the \( (p+\rho_j) \)-th order LPM still requires an initial bandwidth parameter, which is called the pilot bandwidth \( h^* \) and its selection will be discussed later.

Secondly, suppose local homoscedasticity, the estimation covariance of \( \hat{\beta}(t_0) \) can be estimated as:

\[
\hat{\Sigma}(t_0) = (X^T W X)^{-1} \hat{\Sigma}(X^T W X)^{-1} \hat{\Sigma} = (X^T W X)^{-1} \hat{\Sigma} = (X^T W X)^{-1} \hat{\Sigma} = (X^T W X)^{-1} \hat{\Sigma},
\]

The noise variance \( \hat{\sigma}(t_0) \) can be obtained as the normalized weighted residual sum of squares [3]:

\[
\hat{\sigma}(t_0) = \frac{1}{n} \sum_{i=0}^{p+\rho_j} \hat{S}_{kj}(0)p! s(t_k T_i),
\]

where \( X^\text{ex} \) and \( W^\text{ex} \) are respectively the design matrix and weighting matrix in the \( (p+\rho_j) \)-th order LPM using the pilot bandwidth \( h^* \). Consequently, the MSE of \( \hat{\beta}(t_0) \) is

\[
\text{MSE}(\hat{\beta}(t_0)) = \hat{b}_j(t_0) h_j^* + \text{tr}(\hat{\Sigma}(t_0)) t_{tr} \hat{\Sigma}(t_0),
\]

and the optimal bandwidth is determined as the bandwidth having the minimum MSE.

To select a pilot bandwidth \( h^* \), the intersection of confidence interval (ICI) method can be used [5]-[7]. The ICI method selects bandwidth \( h^*(k, t_0) \) for each \( k \)-th coefficient at each time instant \( t \) by calculating the variance of (10) using different bandwidths in \( H \). Because the TVAR coefficients at
each time should be estimated as a whole (using one bandwidth), not individually (using different bandwidths for different k). $h_{uc}(k)$ will be combined (generally, averaged) to produce the pilot bandwidth $h^*(t)$. The algorithm of ICI is omitted to save space, and more details can be found in [5]-[7].

IV. LPM-BASED TVAR FOR EEG

We now apply the LPM-based TVAR identification method for TFA of event-related EEG. By modelling an EEG signal by the TVAR model of (1), we can obtain its time-varying PSD from the estimated coefficients and noise variance as:

$$P(t, f) = \hat{\sigma}^2(t) \left| 1 - \sum_{k=1}^{\infty} \hat{a}_k(t) e^{-j2\pi f} \right|^2.$$  

(13)

We now discuss the parameter selection when applying the LPM method in our EEG study. Following the literature [3]-[7], the Epanechnikov kernel is employed due to its low complexity. The effective size, i.e., the time interval, of an Epanechnikov kernel is $2h$. Next, to determine the bandwidth set $H$ of (8), $h_0$ should be selected as the minimum bandwidth that makes (5) solvable (overdetermined). Because a $(p+p_{ex})$th order LPM is used to approximate MSE, the number of samples included in the kernel should be larger than $(p+p_{ex}+1)L$ (the dimension of $\hat{\mu}(t)$). If the data are uniformly distributed at a sampling frequency of $f_s$, it is required that

$$h_0 \geq (p + p_{ex} + 1)L/(2f_s).$$  

(14)

Moreover, to achieve a good tradeoff between performance and complexity, in our study, the largest bandwidth of $H$ is selected to make the largest kernel cover at least 1/8 of the data set, and the number of bandwidths, $J$, is set as 4. As for the selection of polynomial order $p$ and $p_{ex}$, since the major advantage of the adaptive bandwidth selection is its fine time-resolution, $p$ and $p_{ex}$ should be as small as possible to fulfill the condition of (14). Hence, we set $p=0$ and $p_{ex}=1$ in this study. The proposed bandwidth setting gave satisfactory results in our experiments.

V. EXPERIMENTAL RESULTS AND DISCUSSION

A. Simulated TVAR Model

We first used a second-order TVAR model to illustrate the good performance of the LPM method. The total number of samples is 250 and the additive noise is zero mean with a variance of 0.01. The coefficient $a_1$ jumps from -1.9 to -1.0 at time 125, as shown in Figure 1, while $a_2$ equals a constant -1.0. For comparison, the KF and BEM methods are also tested. In KF, the state transition matrix is chosen as identity matrices to produce a random-walk KF. As mentioned in [1], [11], the ratio between the variances of the state noise and the observation noise, $R_e$ determines the tracking performance of KF. Here, two values (0.01, 1) of $R_e$ are tested in KF. In BEM, Legendre polynomial functions are employed and the number of basis functions is 5.

One simulation is presented in Figure 1. It can be seen that, BEM gives a smooth estimation, but rapid changes are smoothed out at the same time. The performance of KF heavily depends on the selection of $R_e$. A large $R_e$ enables the KF to track fast-varying coefficients while a small $R_e$ helps to reduce the variance of slowly-varying coefficients. In LPM, a small bandwidth can detect fast change of coefficients, but it may lead to large estimation variations, and a large bandwidth can obtain smooth estimates when coefficients vary slowly, but it cannot accurately estimate fast-varying coefficients. LPM with variable bandwidth can obtain satisfactory results for the whole data because of the adaptive variable bandwidths used. The mean squared deviation (MSD) values, $MSD = 10\log_{10} \left\{ \frac{1}{N} \sum_{t=1}^{N} \left( \frac{1}{N} \sum_{k=1}^{N} a(t) - \hat{a}(t) \right)^2 \right\}$, which are obtained by averaging from 100 Monte-Carlo simulations, are given in notation of Figure 1, and we can see that the LPM with variable bandwidth has the minimum MSD value.

B. TFA of Simulated Signals

We next compare various TFA methods, including short-time Fourier transform (STFT), WT and TVAR-based PSD with different identification methods, using a chirp signal with an instantaneous frequency $w(t)$ shown in Figure 2 as red lines. The number of samples is 250 and the sampling frequency is 250Hz. An additive Gaussian white noise with a SNR of 10 dB is added. Hann windows with sizes 20 and 100 are used in STFT, while the Morlet wavelet is used in WT. The order of TVAR models is selected as 4 according to the Akaike’s information criterion. Parameters for TVAR identification methods are the same as those in previous simulation. One representative simulation is presented in Figure 2. The MSD values of estimated instantaneous frequency $\hat{w}(t)$, $MSD = 10\log_{10} \left\{ \frac{1}{N} \sum_{t=1}^{N} \left( w(t) - \hat{w}(t) \right)^2 \right\}$, are obtained by averaging from 100 Monte-Carlo simulations and are listed in notation of Figure 2. It can be seen that the TFD obtained using the LPM with variable bandwidth has better time-frequency resolution than the other methods tested and it can attain fine frequency resolution in a wide frequency range.

Figure 1. Comparisons between various TVAR identification methods: (a) KF with different $R_e$ and BEM, (b) LPM with a constant small kernel ($h_{ex}$), and a constant large kernel ($h_{ex}$), and adaptive variable bandwidths ($h_{ex}$). The MSD values from 100 Monte-Carlo simulations are: KF ($R_e$=1): -16.9 dB; KF ($R_e$=0.01): -16.5 dB; BEM: -15.5 dB; LPM ($h_{ex}$): -17.9 dB; LPM ($h_{ex}$): -16.8 dB; LPM ($h_{ex}$): -20.2 dB.

Figure 2. Comparisons between various TFA methods. The red lines are true instantaneous frequency, and the blue lines are estimated instantaneous frequency (the frequency with the maximum PSD value at each time point). The MSDs of instantaneous frequency estimates from 100 Monte-Carlo simulations are respectively: (a) 11.7 dB; (b) 13.7 dB; (c) 10.2 dB; (d) 19.3 dB; (e) 25.2 dB; (f) 11.2 dB; (g) 16.4 dB; (h): 10.8 dB; (i): 9.5 dB.
C. Real Event-related EEG

A set of visual oddball event-related EEG was collected from the Electrical Geodesic 128-channel EEG system, with vertex referenced. The standard stimulus is a visual character ‘O’ with 80% occurrence, while the target stimulus is a visual character ‘X’ with 20% occurrence. Inter-stimulus interval was 1500ms and the duration of one trial was 1000 ms (100 ms before and 900 ms after stimulus onset). Twelve subjects were recruited and they were required to press the correct button immediately after he/she recognized the stimulus. A total of 250 stimuli were presented, and the sampling rate was 250 Hz. Automatic artifact correction for EEG by blind source separation was performed to attenuate the artifacts [14]. The behavioral data, such as onset time, subject response time and subject response accuracy, were also recorded. The order of TVAR models is selected as 12 by the Akaike's information criterion, which is consistent with previous study [11].

Figure 3 (a) shows the conventional WT- and LPM-based TFDs of an averaged target-response EEG (averaged over 50 trials; recorded at O2) of one subject. It can be seen that, compared with WT, the LPM-based TFD can achieve a higher frequency resolution for high-frequency components and a higher time resolution for low-frequency components. The power peaks of low-frequency components A and B in alpha and theta bands can be seen clearly in LPM-based TFD, and they should be due to the P300 and other early ERP components [8], [9]. A 60-Hz component C starting from 0ms is pronounced in the LPM-based TFD, which should be the steady-state visual evoked potential (SSVEP) in response to computer monitor flicker [15]. In addition, we can see clearly a beta-band oscillation (17-20Hz, component D) after 700ms, which should be the visual attention waiting for the next stimulus [9], [16]. For single-trial EEG, LPM can also achieve a meaningful result. Figure 3 (b) illustrates the TFDs of a single-trial target EEG signal when the subject pressed the wrong button. We can see that, except the 60-Hz SSVEP, P300, and post-response beta oscillation, the component E around 600ms and theta band is distinct. This component should be linked to the error-related negativity (ERN), which occurs when the subject knows his/her error immediately after the response [9]. On the other hand, the ERN pattern in WT is very vague due to the coarse resolution of WT in low frequency bands.

Due to page limitation, more results (different subject and different electrodes) are not presented here. It should be noted that LPM is a powerful method to discover fine time-frequency components of EEG. It facilitates the further analysis, such as statistical analysis, source location, etc., so as to interpret more thoroughly the disclosed time-frequency components and reveal the underlying physiological and psychological mechanisms.

VI. CONCLUSION

A novel LPM approach and a data-driven variable bandwidth selection scheme for identification of TVAR processes were presented. Simulation results showed that the performance of the LPM with adaptive bandwidth selection method is superior to conventional TVAR identification methods in a variety of testing conditions. The proposed method was further applied to TFA of event-related EEG, and experimental results showed that the LPM method can reveal more clearly time-frequency components than WT.

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