RECURSIVE BAYESIAN AUTOREGRESSIVE CHANGE-POINT DETECTOR FOR SEQUENTIAL SIGNAL SEGMENTATION

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ABSTRACT

The contribution addresses a sliding window modification of the Bayesian autoregressive change-point detector (BACD) enabling the sequential localization of signal changes (change-point detection). The modification consists in using the simplified data-dependent Bayesian evidence normalizing the classical BACD formula and in the recursive evaluation of these two functions. The suggested approach seems to be computationally effective and numerical stable as shown by experiments. Apart from the evaluation of the algorithm accuracy two illustrative examples with modelled signals are given. One application to the violin signal segmentation demonstrates the algorithm performance – even relatively weak and gradual signal changes can be detected.

1. INTRODUCTION

The signal segmentation has been intensively studied for decades. A great number of methods and applications have been published. Robust algorithms using recursive identification, likelihood and Bayesian approaches have been designed for the batch or sequential detection of multiple change-points [1], [2], [4], [6].

The first step in signal segmentation is the change-point detection. One of the methods is the Bayesian autoregressive change-point detector [2] which assumes piecewise constant parameters of AR signal and one change-point in one signal segment. The numerical behaviour of the BACD strongly depends on a signal length (number of samples). Greater number of samples gives more consistent estimation of correlations used in the BACD formula but on the other hand causes numerical difficulties. The condition requiring only one change in one signal segment is too limiting for the analysis of real signals as speech, music or biological signals. The segmentation of these signals requires the sequential detection of multiple change-points. Therefore the recursive BACD algorithm based on the sequential Bayesian approach has been suggested [8]. This growing window algorithm uses the data and change-point position updates [2], [7]. Another very simple idea for the sequential change-point detection seems to be the signal segmentation and repeatedly using the BACD. This method requires a proper choice of the segment length and suffers from the sensitivity to a noise. Moreover the BACD results gained from different segments can not be compared. Nevertheless if a normalization of the BACD is used together with the data and change-point position updates then a simple sliding window algorithm suitable for the sequential change-point detection can be obtained.

2. METHOD DESCRIPTION

Firstly, the brief definition of the BACD will be given. Secondly, the BACD use for a segmented signal will be analysed. Finally, the BACD normalization and the recursive algorithm together with the decreasing of computational costs will be described.

2.1 BACD definition

The signal model for the BACD consists of two parts: “left” generated by AR model with \( M_1 \) parameters \( a_k \) and “right” generated by another AR model with \( M_2 \) parameters \( b_k \) [2].

\[
d[n] = \sum_{k=1}^{m} a_k \cdot d[n-k] + e[n], \quad n \leq m
\]

\[
d[n] = \sum_{k=m}^{N} b_k \cdot d[n-k] + e[n], \quad n > m
\]

In matrix form: \( \mathbf{d} = \mathbf{G}_A \cdot \mathbf{b} + \mathbf{e} \). The matrix \( \mathbf{G}_A \) has the Jordan form and depends on the unknown change-point index \( m = 1, \ldots, N \) determined by the maximum of the posterior (MAP) probability density function (pdf) [2].

\[
p(m | \mathbf{d}, M) \propto \left( \frac{D - \mathbf{g}_A \mathbf{r}_A}{\alpha} \right)^{N-M} \cdot \left( \frac{\mathbf{A}_A}{\sqrt{\alpha}} \right)^{M_1 + M_2}, \quad M = M_1 + M_2.
\]

Matrix \( \Phi_A = (\mathbf{G}_A^T \mathbf{G}_A)^{-1} \) is the inverse correlation matrix, \( D = \mathbf{d}^T \mathbf{d} \) is the signal energy, \( \mathbf{g}_A = \mathbf{d}^T \mathbf{G}_A \) is the correlation vector, and \( \Delta_A = \det(\mathbf{G}_A^T \mathbf{G}_A) \). Equation (2) giving the posterior pdf is referred to as the classical BACD formula.

2.2 Signal segmentation and the BACD normalization

Let us consider the problem of the sequential localization of change-points. Starting from the simple idea of repeatedly using the classical BACD on signal segments the BACD modification can be explained. First, the following problems connected with using the classical BACD for the segmented signal will be discussed.
P1. Choice of the segment length. The longer segments are the more consistent estimation of correlations $\Phi$ and $g$ can be obtained but the occurrence more than change-point in one segment is very probably.

P2. Segments overlap. The greater overlapping is the better time resolution can be achieved then but also the greater number of operations is needed.

P3. Posterior pdf normalization. The most important disadvantage of the repeatedly used BACD formula is the impossibility to compare posterior pdfs $p_1(m \mid d, M)$ and $p_2(m \mid d, M)$ (2) belonging to different segments $l, k = 1,2,\ldots,L$. When the BACD is repeatedly used then many candidates for one change-point are generated, and thus a logical filtration of these candidates is required [10] not to obtain false change-points.

Problem P1 cannot be generally solved because it is strongly dependent on the application (signal characteristics). The contradictory requirements P2 can be solved using the recursive evaluation of posterior pdf (2). The sliding recursion decreases the computational costs while preserving the high time resolution (because it uses the overlap $N-1$ samples for segment length $N$ samples). The effective solution of problem P3 is included in the marginalization process leading to pdf (2). As pointed out in [2] and [8] the posterior pdf (2) is influenced only by the nominator of Bayesian formula because it is derived under the condition that the given data segment $d$ is constant [2]. Thus the Bayesian evidence in the denominator of Bayesian formula is constant. But if the pdf (2) is repeatedly used for the segmented signal then the data are not constant. Therefore a modified Bayesian evidence (slightly differing from the Bayesian evidence in [2])

$$BE \approx \frac{\left[D - g \Phi g^T \right]^{-\frac{N-M}{2}}}{\sqrt{\Lambda}}$$

has to be used to normalize the posterior pdf (2). Matrix $G$ is composed of all segment samples without the division into “left” and “right” part. Data and position update (2) includes the update of the change-point

$$p_1(m \mid d, M), \ l = 1,\ldots,L, m = 1,\ldots,N$$

represents the sliding window evaluation of the normalized posterior density (4) for $L$ successive segments giving $N \times L$ values. This great number of values can be further reduced. Instead of using all $N \times L$ samples of function $\hat{p}_1(m \mid d, M)$ the $L$ values can be used only. The information needed for the localization of a change-point is included in time-progress of any function $\hat{p}_1(m \mid d, M), \ m = \text{const.}$ The best choice is $m = N / 2$ giving the sequence $\hat{p}_1(N / 2 \mid d, M), \ l = 1,\ldots,L$ and ensuring the best numerical stability and the lowest sensitivity to disturbances comparing with other asymmetric choices $m \neq N / 2$. It also gives the best results for the model order selection due to fact that “left” and “right” parts of data vector have the same lengths. The evaluation of (4) giving one value of pdf (4) for each segment $\hat{p}_1(m \mid d, M), \ l = 1,\ldots,L, \ m = \text{const.}$ can be also seen as the construction of the longitudinal profile of the “pdgram” (like spectrogram) created by plotting $N$ values of $\hat{p}_1(m \mid d, M), \ l = \text{const.}$ against the segment indexes $l = 1,\ldots,L$. The profile (at $m = N / 2$) of the pdgram is used for the change-point localization then.

2.4 Further simplification of RBACDN

Let $\tilde{p}_1(m \mid d, M), \ l = 1,\ldots,L, m = 1,\ldots,N$ represents the sliding window evaluation of the normalized posterior density (4) for $L$ successive segments giving $N \times L$ values. This great number of values can be further reduced. Instead of using all $N \times L$ samples of function $\hat{p}_1(m \mid d, M)$ the $L$ values can be used only. The information needed for the localization of a change-point is included in time-progress of any function $\tilde{p}_1(m \mid d, M), \ m = \text{const.}$ The best choice is $m = N / 2$ giving the sequence $\tilde{p}_1(N / 2 \mid d, M), \ l = 1,\ldots,L$ and ensuring the best numerical stability and the lowest sensitivity to disturbances comparing with other asymmetric choices $m \neq N / 2$. It also gives the best results for the model order selection due to fact that “left” and “right” parts of data vector have the same lengths. The evaluation of (4) giving one value of pdf (4) for each segment $\tilde{p}_1(m \mid d, M), \ l = 1,\ldots,L, \ m = \text{const.}$ can be also seen as the construction of the longitudinal profile of the “pdgram” (like spectrogram) created by plotting $N$ values of $\tilde{p}_1(m \mid d, M), \ l = \text{const.}$ against the segment indexes $l = 1,\ldots,L$. The profile (at $m = N / 2$) of the pdgram is used for the change-point localization then.

2.5 Summary of RBACDN algorithm

The RBACDN algorithm can be summarized as follows:

I. Initialization of $D, \Phi, \Phi, g, g, \Delta, \Delta$ for $l = 1$ (first data segment) and $m = N / 2 \rightarrow \tilde{p}_1(N / 2 \mid d, M)$

II. Recursion=

\begin{itemize}
  \item Data update for a new sample for (2), (3)
  \item Removing old sample for (2), (3)
  \item Position update for (2)
\end{itemize}

\[
\rightarrow \text{pdgram profile } \tilde{p}_1(N / 2 \mid d, M), l = 2,\ldots
\]

III. Post-processing of $\tilde{p}_1(N / 2 \mid d, M), l = 2,\ldots$

\begin{itemize}
  \item Searching for two adjacent minima of smoothed $\tilde{p}_1(N / 2 \mid d, M)$ (low-pass filter, cut-off $\approx \pi / 100$) \rightarrow stationary points.
  \item Searching for one maximum of $\tilde{p}_1(N / 2 \mid d, M)$ between two stationary points \rightarrow final change-points.
\end{itemize}
Notes on the algorithm: Searching for minima is relatively simple because of their comparable values (close to zero) due to normalization of the BACD by the Bayesian evidence. It can be shown that the minima represent stationary parts of signal. Thus the localization of minima is more robust than the direct localization of maxima. Searching for one maximum of unsmoothed $\tilde{p}_s(N/2 \mid d, M)$ between two minima decreases the number of false alarms and reduces errors. Using the recursive Bayesian evidence also enables to implement an automatic model order selection. In the algorithm representation the logarithm of equation (4) is used from the numerical model order selection. In the algorithm implementation the similarity of the RBACDN and BACD behaviour illustrates figure 1. It can be seen that the MAP of the BACD is at the same position as the maximum of $\tilde{p}_s(N/2 \mid d, M)$. Also shapes of both functions are similar.

Figure 2 shows what happens if the BACD is used for signal with three change-points. This case occurs for real non-stationary signals if the window length is greater than the distance between change-points. While the BACD gives one MAP at index 1400 (or two maxima at 1000 and 1400) the RBACDN generates function $\tilde{p}_s(N/2 \mid d, M)$ with three sharp maxima. When the right part of sliding window reaches the discontinuity in signal (at index 1000) then the left part of the window generates small discontinuity at the same position as the maximum of $\tilde{p}_s(N/2 \mid d, M)$, std=18).

Simulations show relatively low sensitivity of the RBACDN to the model order selection and the high sensitivity to the window length and signal-to-noise ratio. Used window should be longer than 200 samples, otherwise inconsistent results are obtained. For any window longer than 400 samples the RBACDN outperforms the BACD owing to the symmetric matrix $(G_A^T G_A)$. The RBACDN detects changes reliable when the cepstral distance greater than 2 dB.

The similarity of the RBACDN and BACD behaviour illustrates figure 1. It can be seen that the MAP of the BACD is at the same position as the maximum of $\tilde{p}_s(N/2 \mid d, M)$. Also shapes of both functions are similar.

3. EXPERIMENTS AND RESULTS

To validate the RBACDN algorithm the Monte Carlo simulations were used. Various model orders, parameter sets and window lengths were tested. Simulations confirmed that the performances of the RBACDN and the BACD for the detection of one change-point are comparable.

For example: let us consider the signal composed of two 8-order AR processes with equal lengths 4000 samples giving the small change (cepstral distance [11] is about 3 dB - these changes are almost indistinguishable for acoustic signals). Then the RBACDN with window lengths greater than 200 samples gives the average error $|\tilde{m} - m| \approx 5$ samples and standard deviation (std) 20 samples. These values are influenced by the improper change-point estimations falling outside the sliding window. Excluding these errors from the evaluation the RBACDN accuracy is comparable with the BACD accuracy ($|\tilde{m} - m| \approx 3$, std=18).

Next example of segmentation using the RBACDN for real music signal is shown in figure 3 (window length 2000 samples; model orders 8). Because of the continual sliding bow movement on strings the signal does not contain any pauses, signal changes are weak (cepstral distances between segments are in the range from 1.8 to 5 dB) and gradual. Extracted segments bordered by change-points were validated by the multiple model algorithm [12], by the inspection of segmented spectrogram (see fig. 3 bottom), and by listening the isolated segments containing tones.
4. CONCLUSIONS

The sliding window autoregressive change-point detector was suggested and its properties were discussed and illustrated on synthetic and real signals. Further research will be focused on the implementation of the model order selection into the suggested algorithm.

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