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Fast Constrained Independent Component Analysis for Blind Speech Separation with Multiple References

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Abstract- In previous work, the constrained independent component analysis (cICA) algorithm has been proposed to extract the interested signals from the mixtures of some source signals. However, the simultaneous extraction of all signals at the same time presented by cICA prolongs the processing time of this algorithm to extract output signals. In this paper, we introduce a new version of the cICA algorithm to improve cICA in the computational time aspect. By whitening input signals, normalizing weight vectors, and using the one-by-one extraction of output signals, our proposed cICA algorithm has reduced the computational time to recover original signals when compared with the conventional cICA. Meanwhile our proposed cICA algorithm still retains the same recovering performance with that of the conventional cICA. Moreover, in this paper, we also introduce a potential application of our proposed cICA and the conventional cICA on the speech separation problem using priori information to extract the interested speech signals from mixed signals.

I. INTRODUCTION

Blind source separation (BSS) is defined as a method of estimating the original signals from a set of observations that are the mixtures of original signals. In general, a mixing matrix of the original signals is unknown in advanced. A particular example of the BSS problem is the cocktail-party problem depicted in Fig. 1. One might have some speakers in a room, with some microphones used to record the speech signals from the speakers. The task of BSS is to recover the unmixed speech signals of the speakers from the mixed signals received by the microphones.

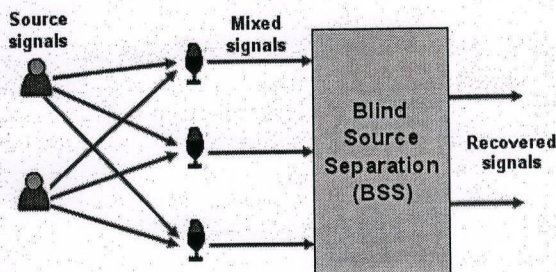


Fig. 1. Blind Source Separation with the cocktail-party problem.

Independent Component Analysis (ICA) [1] is one of the most successful techniques that has been proposed to solve the BSS problem. The idea of ICA is based on the central limit theorem saying that the probability distribution of the mixtures of statistically independent signals tends to follow the Gaussian distribution. Therefore, ICA attempts to extract

independent components (ICs) by finding a demixing matrix of observed signals that maximizes the non-Gaussianity of the extracted signals. The extracted signals become statistically independent and close to the original signals.

Applications of the ICA technique are demonstrated in a large number of areas such as biomedical signal processing, computer vision, or speech processing. In biomedical signal processing, ICA has been popular with extracting the original signals from bio-signal data such as electroencephalogram (EEG) or magnetoencephalogram (MEG) [2][3]. ICA can also be found in the human-machine interaction areas to extract the P300 evoked potential emitted from the human brain to control electronic devices [4]. In computer vision areas, ICA is applied to find the texture information for content based image retrieval [5] or to find a set of basic components of face images for face recognition [6]. In the speech signal processing, ICA is used for speech separation [7][8].

The main existing disadvantage of the ICA algorithm is that the number of recovered signals should be equal to the number of mixed signals. When applying ICA for the incomplete ICA problem where the number of extracted signals is less than the number of mixed signals, the extracted signals are changed over time. The reason causing this problem is that the ICA algorithm only extracts the output signals based on the non-Gaussianity criteria, without using extra information to determine the output signals of interest. On some applications, there might be a lot number of extracted signals but we might be interested in a small number of signals. The rest of extracted signals might be noisy or unmeaningful. In previous work [9], Lu et al. proposed an approach to use cICA to tackle this problem. Reference signals are added to the conventional ICA to drive the extracted signals. This improvement aims at avoiding the arbitrary results of extracting signals and only recovering the signals of interest. However, there are still some drawbacks that lead to the limitations of the cICA algorithm in the computational time aspect: The cICA algorithm attempts to simultaneously recover and decorrelate all output signals at the same time. The cICA algorithm does not consider whitening input signals and normalizing the demixing matrix to restrict the variance values of the output signals when the variances of the original signals are too far from that of the reference signals.

In this paper, we propose a fast version of the cICA algorithm. The fast cICA algorithm achieves faster processing speed by using the one-by-one extraction process of output signals rather than using the simultaneous extraction. The algorithm extracts only one signal at each time, and then

consider the decorrelation constraints between an extracted signal with some previous extracted signals rather than original signals to reduce the computations. Second, the fast cICA algorithm uses the preprocessing with whitening input signals and normalizing weight vectors to bound the variance of output signals that make sure a faster convergence of the cICA algorithm. In addition, in this work, we introduce an improvement of our fast cICA algorithm and the conventional cICA to extract the speech signals of interest from the mixture of speech signals.

This paper is organized as follows. In Section 2, we review the ICA algorithm and summarize the essentials of the cICA algorithm. Our fast cICA algorithm is presented in Section 3. The experimental results are provided in Section 4. Finally, we present our conclusion in Section 5.

2. CONSTRAINED INDEPENDENT COMPONENT ANALYSIS

2.1 Independent component analysis

Consider a BSS problem with n recorders receiving the mixed signals from m different sources. The ICA algorithm attempts to recover m original signals from the n recorded signals without the knowledge about a mixing matrix of the original signals (assuming that the observed signals are the linear mixtures of the original signals). The observed signals $\mathbf{x}(t) = (x_1(t), x_2(t), \dots, x_n(t))^T$ are presented by

$$\mathbf{x}(t) = \mathbf{A}\mathbf{s}(t), \quad (1)$$

where \mathbf{A} is a mixing matrix with size $(n \times m)$ and $\mathbf{s}(t) = (s_1(t), s_2(t), \dots, s_m(t))^T$ is the original signal. The ICA algorithm attempts to compute the demixing matrix $\mathbf{W} = [\mathbf{w}_1, \mathbf{w}_2, \dots, \mathbf{w}_m]^T$ with size $(m \times n)$ to inversely recover the original signals from the observations $\mathbf{x}(t)$

$$\mathbf{y}(t) = \mathbf{W}\mathbf{x}(t), \quad (2)$$

where $\mathbf{y}(t) = (y_1(t), y_2(t), \dots, y_m(t))^T$ is the extracted signal, with $y_i(t) = \mathbf{w}_i^T \mathbf{x}(t)$. Because the extracted signals $y_1(t), y_2(t), \dots, y_m(t)$ are close to the original sources, they should be mutually independent. The ICA algorithm aims at finding the demixing matrix \mathbf{W} to maximize the non-Gaussianity of the extracted signal $\mathbf{y}(t)$ that causes $\mathbf{y}(t)$ to be mutually independent and converge toward one of the ICs. The measurement of the non-Gaussianity is given by the negentropy $J(\mathbf{y})$ [1]

$$J(\mathbf{y}) \approx \rho[E\{f(\mathbf{y})\} - E\{f(\mathbf{v})\}]^2, \quad (3)$$

where \mathbf{v} is a Gaussian variable with zero mean and unit variance. Some available functions are suggested for f

$$f_1(y) \approx \log \cosh(ay) / a, \quad (4)$$

$$f_2(y) \approx \exp(-ay^2/2) / a, \quad (5)$$

$$f_3(y) \approx y^4 / 4, \quad (6)$$

where a is a positive constant. The function f_1 is mostly used for the general case, f_2 is used for the super-Gaussian signal, and f_3 is used for the sub-Gaussian signal, respectively.

Because the non-Gaussianity is an only criterion used to extract output signals, there will be an arbitrary ordering of extracted ICs. When we want to extract less than the number of observations, the extracted results are changed over time and we cannot recover the signals of interest.

2.2 Constrained independent component analysis

The cICA algorithm is developed to retrieve only desired ICs by using additional constraints to drive the extractions of output signals. We summarize the essentials of cICA in this section. More details of cICA can be found in [9].

The cICA algorithm integrates some equality constraints $h(\mathbf{y}; \mathbf{W})$ and inequality constraints $g(\mathbf{y}; \mathbf{W})$ into the optimization function (3) of the ICA algorithm. The overall optimization equation of cICA is rewritten by

$$\max \sum_{i=1}^m J(y_i) \approx \sum_{i=1}^m \rho[E\{f_i(y_i)\} - E\{f_i(\mathbf{v})\}]^2 \quad (7)$$

$$\text{subject to } h(\mathbf{y}; \mathbf{W}) = 0, g(\mathbf{y}; \mathbf{W}) \leq 0$$

or

$$\min - \sum_{i=1}^m J(y_i) \approx - \sum_{i=1}^m \rho[E\{f_i(y_i)\} - E\{f_i(\mathbf{v})\}]^2 \quad (8)$$

$$\text{subject to } h(\mathbf{y}; \mathbf{W}) = 0, g(\mathbf{y}; \mathbf{W}) \leq 0$$

where $h(\mathbf{y}; \mathbf{W}) = (h_{11}(y_1), h_{12}(y_1, y_2), \dots, h_{mm}(y_m))^T$ and $g(\mathbf{y}; \mathbf{W}) = (g_1(y_1), g_2(y_2), \dots, g_m(y_m))^T$.

A set of equality constraints $h(\mathbf{y}; \mathbf{W})$ to make the output signals uncorrelated and bound the variance values of the output signals to be one is given by

$$h_{ij}(y_i, y_j) = (E\{y_i y_j\})^2 = 0, \forall i, j = 1, 2, \dots, m, i \neq j \quad (9)$$

$$h_{ii}(y_i) = (E\{y_i^2\} - 1)^2 = 0, i = 1, 2, \dots, m. \quad (10)$$

The priority information is provided to the cICA algorithm by some reference signals. A set of inequality constraints $g(\mathbf{y}; \mathbf{W})$ used to make the output signals closed to the reference signals is given by

$$g_i(y_i) = \varepsilon(y_i, r_i) - \xi_i \leq 0, i = 1, 2, \dots, m, \quad (11)$$

where ξ_i is a threshold and $\varepsilon(y_i, r_i)$ is the closeness measurement between y_i and r_i . The common formulation of $\varepsilon(y_i, r_i)$ is the mean squared error $\varepsilon(y_i, r_i) = E\{(y_i - r_i)^2\}$ and the inverse correlation $\varepsilon(y_i, r_i) = 1/E\{y_i r_i\}^2$.

To solve an optimization problem with equality constraints and inequality constraints, we need to add extra variables into an optimization function with the Lagrange multipliers. The expansion of the cICA algorithm with the Lagrange multipliers is completely described in [9]. Here, we only rewrite the update rules to calculate the values of the weight matrix and the Lagrange multipliers by the Newton's method,

$$\mathbf{W} \leftarrow \mathbf{W} - \eta \mathbf{D}^{-1} \left(-E\{\nabla \mathbf{J}(\mathbf{y}) \mathbf{x}^T\} + \Gamma \nabla \mathbf{G} + 2\Lambda E\{\mathbf{y} \mathbf{x}^T\} \right) \Sigma_{xx}^{-1}, \quad (12)$$

$$\mu \leftarrow \max\{0, \mu + \gamma_1 g(\mathbf{y}; \mathbf{W})\}, \quad (13)$$

$$\lambda \leftarrow \lambda + \gamma_2 h(\mathbf{y}; \mathbf{W}), \quad (14)$$

where η , γ_1 and γ_2 are the learning rates, $\mu = [\mu_1, \mu_2, \dots, \mu_m]^T$ and $\lambda = [\lambda_{11}, \lambda_{12}, \dots, \lambda_{mm}]^T$ are the Lagrange multipliers, $\mathbf{D} = \text{diag}(d_1, d_2, \dots, d_m)$ is a diagonal matrix with each diagonal element $d_i = -E\{\hat{\rho}_i f''_{y_i}(y_i)\} + 8\lambda_{ii} + E\{\mu_i g''_{y_i}(y_i)\}$,

$$\nabla \mathbf{J}(\mathbf{y}) = [J'(y_1), J'(y_2), \dots, J'(y_m)]^T, \quad \Gamma = \text{diag}(\mu_1, \mu_2, \dots, \mu_m),$$

$$\nabla \mathbf{G} = [\nabla_{y_1} g_1(y_1), \nabla_{y_2} g_2(y_2), \dots, \nabla_{y_m} g_m(y_m)]^T, \quad \Sigma_{xx} = E\{\mathbf{x} \mathbf{x}^T\},$$

$$\text{and } \Lambda \text{ is a matrix with } \Lambda_{ij} = \begin{cases} \lambda_{ij} E\{y_i y_j\} & i \neq j \\ \lambda_{ii} (E\{y_i^2\} - 1) & i = j \end{cases}. \text{ Here,}$$

we use $\nabla_{y_i} g(y)$ rather than $E\{g'(y) \mathbf{x}^T\}$ as recommended by Liu et al. [10] to make sure that the equality constraints are still valid when $E\{\mathbf{x}^T\}$ is close to 0.

III. FAST CONSTRAINED INDEPENDENT COMPONENT ANALYSIS

A. Fast constrained independent component analysis

In this section, we develop the fast cICA with multiple references by the two modifications on the conventional cICA. First, the preprocessing with whitening input signals and normalizing weight vectors at each update step are integrated into the cICA algorithm, as in [11]. This addition aims at bounding the variance values of extracted signals that make the estimation of the output signals faster. Second, we perform the one-by-one extraction process of the output signals and reduce the complexity of equality constraints in (9), (10), and (11). The simultaneous decorrelation of all output signals at the same time makes cICA be harder to converge to stable values. The mathematical details of the fast cICA algorithm and its update rule to learn the weight vector are presented following.

The gradient of the optimization problem in (8) for one signal with the addition of the Lagrange multipliers is given by

$$\nabla L_w = -E\{J'(\mathbf{y}) \mathbf{x}^T\} + \mu \nabla_{y_p} g(y) + 4\lambda (E\{y^2\} - 1) E\{\mathbf{y} \mathbf{x}^T\}. \quad (15)$$

The last term $4\lambda (E\{y^2\} - 1) E\{\mathbf{y} \mathbf{x}^T\}$ used to restrict the variance of output signal to a value of one is replaced by normalizing the weight vector at each step $\mathbf{w} \leftarrow \mathbf{w} / \|\mathbf{w}\|$. For the output signal p , we need to establish $p-1$ constraints $(E\{y_j y_p\})^2 = 0, \forall j = 1, 2, \dots, p-1$ to decorrelate the output signal p with the $p-1$ extracted signals. According to the Kuhn-Tucker theorem, with these additional constraints, a set of equations used to compute the values of the weight vector is depicted by

$$\begin{cases} \nabla L_{w_p} = -E\{J'(\mathbf{y}) \mathbf{x}^T\} + \mu \nabla_{y_p} g(y) + 2 \sum_{j=1}^{p-1} \lambda_{jp} E\{y_j y_p\} E\{\mathbf{y} \mathbf{x}^T\} = 0 \\ (E\{y_j y_p\})^2 = 0, \forall j, j = 1, 2, \dots, p-1 \end{cases} \quad (16)$$

where λ_{jp} is the Lagrange multiplier. We attempt to solve these equations by the gradient descent with the Newton method. The Jacobian matrix of L_{w_p} is approximated

$$\nabla L_{w_p}^2 = (-E\{\hat{\rho} f''(y_p)\} + E\{\mu g_p''(y_p)\}) \Sigma_{xx} + 2 \sum_{j=1}^{p-1} \lambda_{jp} \mathbf{w}_j \mathbf{w}_j^T,$$

where $\hat{\rho} = 2\rho(E\{f(y_p)\} - E\{f(y)\})$. We perform the approximation by replacing $E\{y_j y_p\} E\{\mathbf{y} \mathbf{x}^T\}$ with $(\mathbf{w}_p^T \mathbf{w}_j) \mathbf{w}_j$.

Finally, we have the update rules to compute each weight vector as below

$$\mathbf{w}_p \leftarrow \mathbf{w}_p - \eta \nabla_{w_p} L^{-1} (-E\{J'(\mathbf{y}) \mathbf{x}^T\} + \mu \nabla_{y_p} g(y))$$

$$+ 2 \sum_{j=1}^{p-1} \lambda_{jp} (\mathbf{w}_p^T \mathbf{w}_j) \mathbf{w}_j,$$

$$\mathbf{w}_p \leftarrow \mathbf{w}_p / \|\mathbf{w}_p\|$$

where η is a learning rate. Here, the weight vector \mathbf{w}_p is initialized with a value from the uniform distribution.

μ are set to zero at the first iteration. The update algorithm is iterated until the weight vector \mathbf{w}_p converges to a stable value.

B. The roles of whitening input signals and normalizing weight vectors for extracting the signals with high variance values

In the conventional cICA algorithm, we need to solve the equality constraints in (9) to make the output signals uncorrelated. However, with the output signals with high variance values, the correlation formulation in (9) needs to be rewritten by

$$h_{ij}(y_i, y_j) = (E\{y_i y_j\})^2 / (E\{y_i^2\} E\{y_j^2\}) = 0, i \neq j,$$

to make $h_{ij}(y_i, y_j)$ converge faster to zero. Moreover,

with $E\{y^2\} \gg 1$, the values of $4\lambda_{ii} (3E\{y_i^2\} - 1)$ can be approximated by $8\lambda_{ii}$ as in d_i of equation (12). The update

learning rule for the weight matrix of the conventional cICA given in (12) needs to have the following adjustments: $\mathbf{D} = \text{diag}(d_1, d_2, \dots, d_m)$ is a diagonal matrix with each diagonal element

$$d_i = -E\{\hat{\rho}_i f''_{y_i}(y_i)\} + 4\lambda_{ii} (3E\{y_i^2\} - 1) + E\{\mu_i g''_{y_i}(y_i)\} + 2 \sum_{j, j \neq i} \lambda_{ij} / E\{y_i^2\},$$

and Λ is a matrix with

$$\Lambda_{ij} = \begin{cases} \lambda_{ij} E\{y_i y_j\} / (E\{y_i^2\} E\{y_j^2\}) & i \neq j \\ \lambda_{ii} (E\{y_i^2\} - 1) - \sum_{j, j \neq i} \lambda_{ij} E\{y_i y_j\}^2 / (E\{y_i^2\}^2 E\{y_j^2\}) & i = j \end{cases}$$

For the fast cICA, it is unnecessary to integrate the denominator $E\{y_i^2\} E\{y_j^2\}$ into the equation $h_{ij}(y_i, y_j)$.

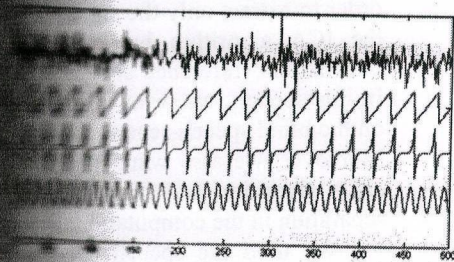
Because the variance $E\{y^2\} = \mathbf{w}^T E\{\mathbf{x} \mathbf{x}^T\} \mathbf{w}$ always gets a value of one, the weight vector is normalized at each update step $\|\mathbf{w}\| = \mathbf{w}^T \mathbf{w} = 1$, and whitening input signals makes Σ_{xx} become an identity matrix.

We present the experimental results in the following section have it with the conventional cICA algorithm. The speed of the learning rules in equation (19) and (20) is approximately 10 times faster than the conventional cICA algorithm. The fast cICA algorithm is still superior to the conventional cICA algorithm and even the cICA algorithm on the processing time aspect. The experimental results with synthetic data and normalizing weight vectors have shown that the fast cICA algorithm is superior to the conventional cICA algorithm in the computations of our fast computational complexity and reduced the computational complexity in the computations of the weight vectors.

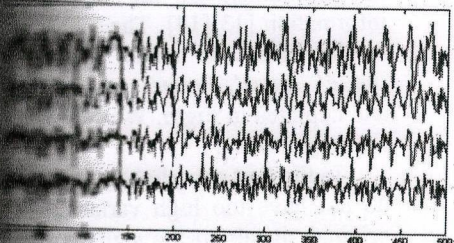
IV. EXPERIMENTS

A. Experiments with synthetic data

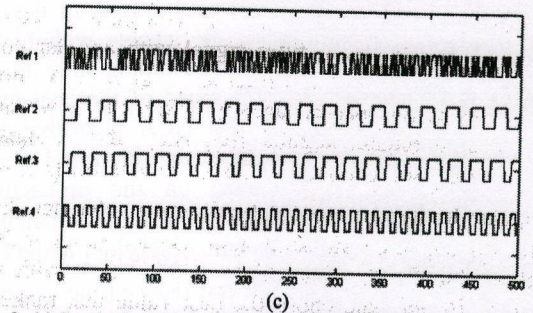
In this section, we present the experimental results with synthetic data to compare the performance of our fast cICA algorithm with the conventional cICA algorithm. The programming codes used for the experiments with synthetic data are provided in the software package. The original signals of the conventional cICA algorithm [12]. The original signals of the fast cICA algorithm are shown in Fig. 2(a). The mixing matrix of the conventional cICA algorithm is created randomly and the reference signals are generated by the sign of the original signals. The mixed signals are depicted in Fig. 2(b) and the reference signals are shown in Fig. 2(c), respectively. To compare the performance of our fast cICA and the conventional cICA algorithm, we compute the peak-signal-to-noise ratio (PSNR) and the mean squared error (MSE) between the extracted signal and the original signal. The PSNR is defined as $10 \log_{10}(\sigma^2 / \text{MSE})$ (σ^2 is the variance of the original signal and MSE is the mean squared error between the extracted signal and the original signal) and the absolute correlation is defined as $|\langle x, y \rangle| / \sqrt{\langle x, x \rangle \langle y, y \rangle}$. The output signals are better than the conventional cICA algorithm. The PSNR values are higher and their absolute correlation values are close to one.



(a)



(b)



(c)

Fig. 2. (a) Synthetic data with the original signals. (b) The mixed signals. (c) The reference signals.

The experimental results with the synthetic data given in Table 1 have shown that our fast cICA takes lesser time to recover the original signals than does the conventional cICA (Note that in this table, 'Src' is used to abbreviate 'Source'). Meanwhile, in the recovering performance aspect, our fast cICA algorithm has similar values of PSNR and absolute correlation with those of the conventional cICA algorithm.

TABLE I
COMPARISON OF RECOVERING PERFORMANCES AND RUNNING TIMES BETWEEN CICA AND FAST CICA ON SYNTHETIC DATA

		Src 1	Src 2	Src 3	Src 4	Running Time (s)
Conventional cICA	Absolute Correlation	1	0.99	1	1	0.34
	PSNR (dB)	32.78	28.22	33.11	25.48	
Fast cICA	Absolute Correlation	1	1	1	0.99	0.18
	PSNR (dB)	34.57	24.95	34.31	28.91	

B. Experiments with speech data

In recent years, there a lot of attempts to apply the ICA algorithm on recovering the original speech signals from a set of mixed signals (the blind speech separation). However, in order to extract only the signals of interest, we need to replace the ICA algorithm by the cICA algorithm to use priori information (reference signals) to drive the extracted signals. In this section, we want to introduce the application of cICA on the blind speech separation with multiple reference signals. Moreover, we also want to test the performance of our fast cICA algorithm and conventional cICA on the blind speech separation in the computational time aspect.

We used the speech dataset in the website storing the implementation of the ICA algorithm based on mutual information [13][14]. The eight speech signals corresponding to the files with name 'alex', 'dave2', 'daver', 'doors', 'halle', 'inter', 'main1', 'main2' are used in our experiments and depicted in Fig. 3. We call these eight signals by S_1, S_2, \dots, S_8 , respectively.

In the first experiment, we assume that the original signals do not have too high variance values, so we normalize the original signals to the signals with unit variance. The mixed signals are generated from the original signals using a random mixing matrix. We want to use our fast cICA algorithm and the conventional cICA to extract the signals S_3, S_4 , and S_8 from the mixed signals. The missing-frequency signals are used as the reference signals for S_3 and S_4 : the high-pass filter signal

with angular cut-off frequency 0.7rad/s is used as a reference signal for S_3 ; the low-pass filter signal with angular cut-off frequency 0.3rad/s is used as a reference signal for S_4 . For S_8 , observing the power spectral density (PSD) of S_8 , we notice that the most dominant angular frequency of this signal is 0.18rad/s . Thus, we choose a sinusoid signal $\cos(\omega(t+t_0))$ with angular frequency $\omega = 0.18\text{rad/s}$ as a reference signal. The time dilate t_0 of the sinusoid signal is set a value of 20. To chose the value for the time dilate t_0 , we test t_0 with some values 0, 5, 10, etc. and chose the best value that makes the convergence of cICA stable. The signals S_3 , S_4 , and S_8 recovered from the mixed signals by the conventional cICA algorithm and the fast cICA algorithm are shown in Fig. 4(a) and 4(b), respectively. From the experimental results given in Table 2, we can see that our fast cICA achieves the same recovering performance as that of the conventional cICA. However, in the running time aspect, the speed of our fast cICA is about three times faster than that of the conventional cICA.

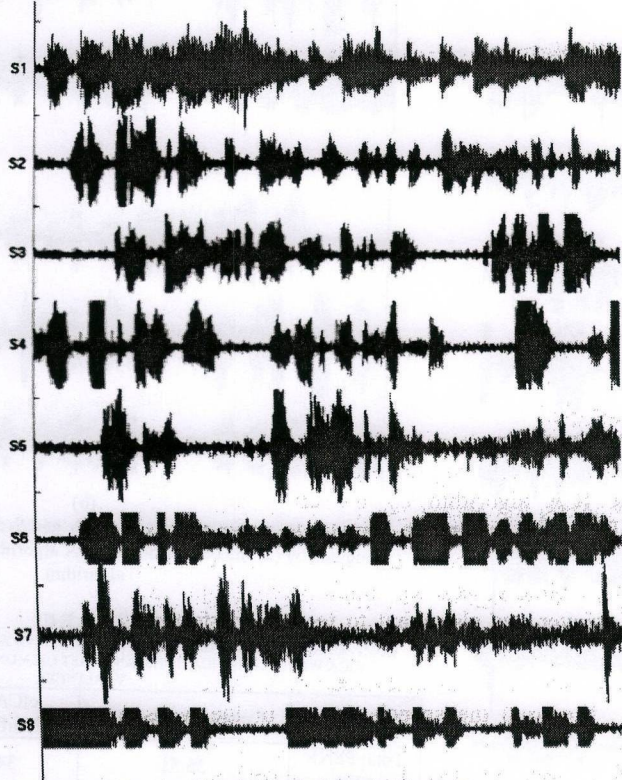


Fig. 3. The original speech signals.

TABLE II
COMPARISON OF RECOVERING PERFORMANCES AND RUNNING TIMES BETWEEN CICA AND FAST CICA ON SPEECH DATA.

		S3	S4	S8	Running Time (s)
Conventional cICA	Absolute Correlation	1.04	0.96	0.94	5.55
	PSNR (dB)	24.44	8.81	8.53	
Fast cICA	Absolute Correlation	1.04	0.97	0.94	1.89
	PSNR (dB)	24.79	8.89	8.56	

In the second experiment, we test the running of cICA and fast cICA on the signals with high variance values. The original speech signals without normalizing variance are used in our experiment. We want to extract the two signals S_4 and S_7 from the mixed signals. The low-pass filter signals S_4 and S_7 with angular cut-off frequency 0.4rad/s are used as the reference signals. In this case, the cICA algorithm requires some adjustments as in Section III.B to make cICA converge faster. However, even with some modifications, cICA consumes more time than our fast cICA to extract the original signals S_4 and S_7 , as shown in Table 3.

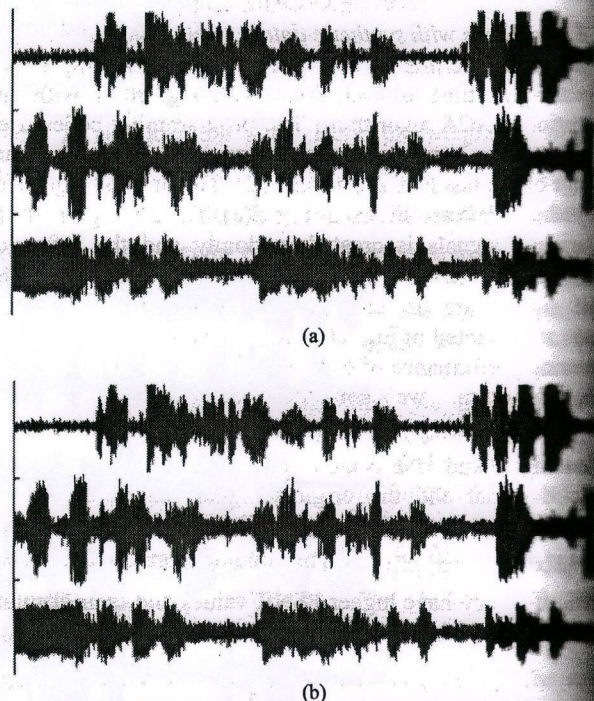


Fig. 4. The interested signals S_3 , S_4 , and S_8 are estimated from the mixed signals by (a) the conventional cICA algorithm and (b) the fast cICA algorithm.

TABLE III
COMPARISON OF RECOVERING PERFORMANCES AND RUNNING TIMES BETWEEN CICA WITH MODIFICATIONS, AND FAST CICA ON SPEECH DATA WITH HIGH VARIANCES.

	Conventional cICA	cICA with Modifications	Fast cICA
Total PSNR (dB)	35.51	34.55	35.51
Running Time (s)	20.44	5.30	1.89

IV. CONCLUSIONS

In this paper, we have proposed a new version of the cICA algorithm to improve this algorithm in the computational aspect. The experimental results with the synthetic speech data have shown that our algorithm runs faster than the conventional algorithm, meanwhile it still maintains the accurate performance of recovering the original signals. Therefore, our algorithm must be better for being applied in real applications.

that the cICA algorithm and the fast cICA algorithm for solving the blind speech separation problem with the aim of extracting only the signals of interest. The successful extraction of specific information of the signals such as amplitude, or low or high frequency components, will allow us to recover the original speech signals from the mixed signals with high accuracy. The successful extraction of speech signals from the mixed signals will be useful in a wide range of applications related to speech processing, such as speech enhancement, and audio editing.

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REFERENCES

1. P. Comon, "Independent Component Analysis," *IEEE Trans. on Neural Networks*, vol. 10, pp. 1300-1302, 1999.
2. A. J. Bell and T. J. Sejnowski, "An information maximization approach to blind separation and blind deconvolution," *Neural Computation*, vol. 7, pp. 1129-1159, 1995.
3. S. S. St. Jacques, J. Karhunen, E. Oja, *Independent Component Analysis*. John Wiley and Sons, 2001.
4. S. S. St. Jacques, C. J. James, T. Nakahori, T. Akiyama, and J. Gotman, "Extraction of epileptiform discharges from unaveraged EEG by independent component analysis," *Electroenceph. Clin. Neurophysiol.*, vol. 109, pp. 1755-1763, 1999.
5. S. S. St. Jacques and D. Lowe, "ICA in electromagnetic brain signal analysis," *Proc. of the Conf. Neural Networks and Expert Systems in Medicine*, vol. 1, pp. 197-202, 2001.
6. S. S. St. Jacques, B. Hong, and S. Gao, "BCI competition 2003-Dataset 2: Improving P300 wave detection using ICA-based subspace analysis for BCI applications," *IEEE Trans. on Biomedical Engineering*, vol. 51, pp. 1067-1072, 2004.
7. S. S. St. Jacques and Y. Matsuyama, "Database retrieval for similar images using ICA and PCA bases," *Engineering Applications of Artificial Intelligence*, vol. 18, pp. 705-717, 2005.
8. S. S. St. Jacques, J. R. Movellan, and T. J. Sejnowski, "Face recognition by independent component analysis," *IEEE Trans. on Neural Networks*, vol. 13, pp. 1450-1464, 2002.
9. S. S. St. Jacques, J. Yin, B. Ayhan, S. Chu, X. Liu, K. Puckett, Y. Zhao, K. C. Frick, and I. Sityar, "Speech separation algorithms for multiple speaker environments," in *Proc. of Int. Joint Conf. on Neural Networks (IJCNN)*, 2008, China, June 2008, pp. 1644-1648.
10. S. S. St. Jacques, S. Makino, H. Sawada, and R. Mukai, "Underdetermined blind speech separation with directivity pattern based continuous mask and ICA," in *Proc. of Int. Conf. ICA 2004*, Spain, September 2004, pp. 1095-1099.
11. S. S. St. Jacques and J. C. Rajapakse, "Approach and applications of constrained ICA," *IEEE Trans. on Neural Networks*, vol. 16, pp. 203-212, 2005.
12. S. S. St. Jacques and J. X. Mi, "A new constrained independent component analysis method," *IEEE Trans. on Neural Networks*, vol. 18, pp. 1532-1545, 2007.
13. S. S. St. Jacques, Y. Zheng, F. Yin, H. Liang, and V. D. Calhoun, "A fast algorithm for the one-unit ICA-R," *Information Sciences*, vol. 177, pp. 1265-1275, 2007.
14. <http://www.cis.hut.fi/projects/ica/fastica/>
15. <http://cnl.salk.edu/pub/tony/sounds/>
16. A. J. Bell and T. J. Sejnowski, "An information maximisation approach to blind separation and blind deconvolution," *Neural Computation*, vol. 7, pp. 1129-1159, 1995.