Hierarchical Probabilistic Model for Blind Source Separation via Legendre Transformation

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Abstract

We present a novel blind source separation (BSS) method, called information geometric blind source separation (IGBSS). Our formulation is based on the log-linear model equipped with a hierarchically structured sample space, which has theoretical guarantees to uniquely recover a set of source signals by minimizing the KL divergence from a set of mixed signals. Source signals, received signals, and mixing matrices are realized as different layers in our hierarchical sample space. Our empirical results have demonstrated on images and time series data that our approach is superior to well established techniques and is able to separate signals with complex interactions.

1 INTRODUCTION

The objective of blind source separation (BSS) is to identify a set of source signals from a set of multivariate mixed signal. BSS is widely used for applications which are considered to be the “cocktail party problem”. Examples include image/signal processing [Isomura and Toyoizumi, 2016], artifact removal in medical imaging [Vigario et al., 1998], and electroencephalogram (EEG) signal separation [Congedo et al., 2008]. Currently, there are a number of solutions for the BSS problem. The most widely used approaches are variations of principal component analysis (PCA) [Pearson, 1901, Murphy, 2012] and independent component analysis (ICA) [Comon, 1994, Murphy, 2012]. However, they all have limitations with their approaches.

PCA and its modern variations such as sparse PCA (SPCA) [Zou et al., 2006], non-linear PCA (NLPCA) [Scholz et al., 2005], and Robust PCA [Xu et al., 2010] extract a specified number of components with the largest variance under an orthogonal constraint. They are composed of a linear combination of variables, and create a set of uncorrelated orthogonal basis vectors that represent the source signal. The basis vectors with the N largest variance are called the principal components and are the output of the model. PCA has shown to be effective for many applications such as dimensionality reduction and feature extraction. However, for BSS, PCA makes the assumption that the source signals are orthogonal, which is often not the case in most practical applications.

Similarly, ICA also attempts to find the N components with the largest variance by relaxing the orthogonality constraint. Variations of ICA, such as infomax [Bell and Sejnowski, 1995], FastICA [Hyvärinen and Oja, 2000], and JADE [Cardoso, 1999], separate a multivariate signal into additive subcomponents by maximizing the statistical independence of each component. ICA assumes that each component is non-gaussian and the relationship between the source signal and the mixed signal is an affine transformation. In addition to these assumptions, ICA is sensitive to the initialization of the weights as the optimization is non-convex and is likely to converge to a local optimum.

Other potential methods which can perform BSS include non-negative matrix factorization (NMF) [Lee and Seung, 2001, Berne et al., 2007], dictionary learning (DL) [Olshausen and Field, 1997], and reconstruction ICA (RICA) [Le et al., 2011]. NMF, DL and RICA are degenerate approaches to recover the source signal from the mixed signal, which means that they lose information when recovering the source signal. These approaches are more typically used for feature extraction. NMF factorizes a matrix into two matrices with nonnegative elements representing weights and features. The features extracted by NMF can be used to recover the source signal. More recently, there are more advanced techniques that uses Short-time Fourier transform (STFT) to transform the signal into the frequency domain to construct a spectrogram before applying NMF [Sawada et al., 2019]. However, NMF does not maximize statistical
independence which is required to completely separate the mixed signal into the source signal, and it is also sensitive to initialization as the optimization is non-convex. Due to the non-convexity, additional constraints or heuristics for weight initialization is often applied to NMF to achieve better results [Ding et al., 2008, Boutsidis and Gallopoulos 2008]. DL can be thought of as a variation of the ICA approaches which requires an over-complete basis vector for the mixing matrix. DL may be advantageous because additional constraints such as a positive code or a dictionary can be applied to the model. However, since it requires an over-complete basis vector, information may be lost when reconstructing the source signal. In addition, like all the other approaches, DL is also non-convex and it is sensitive to the initialization of the weights.

All previous approaches have limitations such as loss of information or non-convex optimization and require constraints or assumptions such as orthogonality or an affine transformation which are not ideal for BSS. In the following, we introduce our approach to BSS, called IGBSS (Information Geometric BSS), using the log-linear model [Agresti 2012], which can introduce relationships between possible states into its sample space [Sugiyama et al. 2017]. Unlike the previous approaches that we mentioned above, our approach does not have the assumptions or limitations that they require. We provide a flexible solution by introducing a hierarchical structure between signals into our model, which allows us to treat interactions between signals that are more complex than an affine transformation. Unlike other existing methods, our approach does not require the inversion of the mixing matrix and is able to recover the sign of the signal. Thanks to the well-developed information geometric analysis of the log-linear model [Amari 2001], optimization of our method is achieved via convex optimization, hence it always arrives at the globally optimal unique solution. We theoretically show that it always minimizes the Kullback–Leibler (KL) divergence from a set of mixed signals to a set of source signals. We empirically demonstrate that our hierarchical model leads to better separation of signals including complex interaction such as higher-order feature interactions than existing methods.

2 FORMULATION

BSS is formulated as a function \( f \) that separates a set of received signals \( X \) into a set of source signals \( Z \), i.e., \( Z = f(X) \). For example, if one employs a ICA based formulation, the BSS problem reduces to \( X = AZ \), where the received signal \( X \in \mathbb{R}^{L \times M} \) with \( L \) signals and the sample size \( M \) is an affine transformation of the source signal \( Z \in \mathbb{R}^{N \times M} \) with \( N \) signals and a mixing matrix \( A \in \mathbb{R}^{L \times N} \). The objective is to estimate \( Z \) by learning \( A \) given \( X \). Our approach is different from the classical formulation, where the inverse of the mixing matrix is learnt to recover the source signal, that is \( Z = A^{-1}X = BX \).

Our strategy is to treat the three components, \( X, Z, \) and \( A \), of BSS as a joint distribution and model it by the log-linear model [Agresti 2012], which is a well-known energy-based model. We can take non-affine transformation into account and formulate BSS as a convex optimization problem.

\[
\begin{bmatrix}
a_{11} & a_{12} \\
a_{21} & a_{22}
\end{bmatrix}
\begin{bmatrix}
z_{11} & z_{12} \\
z_{21} & z_{22}
\end{bmatrix} =
\begin{bmatrix}
x_{11} & x_{12} \\
x_{21} & x_{22}
\end{bmatrix}
\]

\begin{figure}[h]
\centering
\includegraphics[width=0.5\textwidth]{sample_space.png}
\caption{An example of our sample space. Dashed lines show removed partial orders to allow for learning. The nodes represent the state of each variable and the arrow shows the direction of the partial ordering.}
\end{figure}

2.1 LAYER CONFIGURATION

Let \( \Omega \) be a sample space of distributions modeled by the log-linear model, which is composed of possible states of a system of interest. Our key idea is to introduce a hierarchical layered structure into \( \Omega \) to achieve BSS. We call this model information geometric BSS (IGBSS) as its optimality is supported by the tight connection between the log-linear model and the information geometric properties of the space of distributions (statistical manifold), which we will show in the following subsections. We implement three layers of BSS, the mixing layer, the source layer, and the received layer, into \( \Omega \) in the form of partial orders and learn the joint representation on it using the log-linear model. The log-linear model on a partially ordered set (poset), a set equipped with a partial order \( \preceq \) [Gierz et al. 2003], is proposed by Sugiyama et al. [2017], which includes a (higher-order) Boltzmann machines as an instance [Luo and Sugiyama 2019]. We use this model to achieve the task of BSS by introducing layered structure as partial orders. The received layer and the source layer represent the input received signal and the output source signal of BSS, respectively, and the mixing layer encodes information of how to mix the source signal. In the following, we consistently assume that \( L \) is the number of received signals, \( M \) is the sample size, and \( N \) is the number of source signals.

Let us construct three layers in the sample space \( \Omega \) as:
\[ \Omega = \{ \perp \} \cup A \cup Z \cup \mathcal{X} \] and assume that these sets are given as \( A = \{ a_{11}, \ldots, a_{LM} \} \), \( Z = \{ z_{11}, \ldots, z_{NM} \} \), and \( \mathcal{X} = \{ x_{11}, \ldots, x_{LM} \} \). The element \( \perp \) denotes the least element, and it acts as a partition function of the log-linear model. We use 2D indexing of elements in each layer to make the correspondence between our formulation and ICA based formulation clear; that is, these three layers, \( A \), \( Z \), and \( \mathcal{X} \) are analogous to a mixing matrix \( A \in \mathbb{R}^{L \times N} \), a source matrix \( Z \in \mathbb{R}^{N \times M} \), and a received matrix \( X \in \mathbb{R}^{L \times M} \), respectively.\(^2\) We will also use symbols \( \omega \) and \( s \) to denote elements of \( \Omega \), i.e., they can be \( \perp, a_{lm}, z_{mn}, \) and \( x_{lm} \). Here we introduce a partial order \( \preceq \) between layers. We introduce the connection between each of the nodes in a similar fashion to the forward ICA model, \( X = AZ \). We define

\[
\begin{align*}
   a_{ij} &\preceq z_{i'j'} \quad \text{if } j = i', \\
   a_{ij} &\not\preceq z_{i'j'} \quad \text{otherwise}, \\
   z_{ij} &\preceq x_{i'j'} \quad \text{if } j = j', \\
   z_{ij} &\not\preceq x_{i'j'} \quad \text{otherwise} \quad (1)
\end{align*}
\]

for each element in three layers, \( A \), \( Z \), and \( \mathcal{X} \), and we do not have any ordering among elements in the same layer. Since it is a partial order, transitivity always holds, e.g., \( a_{11} \preceq x_{22} \) as \( a_{11} \preceq z_{12} \) and \( z_{12} \preceq x_{22} \). The first condition encodes the structure such that the source layer is higher than the mixing layer, and the second condition encodes that the received layer is higher than the source layer. An example of our sample space with \( L = M = N = 2 \) is illustrated in Figure 1.

### 2.2 Log-linear Model on Partially Ordered Sets

We use the log-linear model given in the form of

\[
\log p(\omega) = \sum_{\omega \in S} 1_{s \preceq \omega} \theta_s - \psi(\theta),
\]

where \( p(\omega) \in (0, 1) \) is the probability of each state \( \omega \in \Omega \) and \( S \subseteq \Omega \) is a parameter space such that a parameter value \( \theta_s \in \mathbb{R} \) is associated with each \( s \in S \), and \( \psi(\theta) \) is the partition function such that \( \sum_{\omega \in \Omega} p(\omega) = 1 \), where \( \theta_{\perp} = -\psi(\theta) \) always holds. In this formulation, we assume that the set \( \Omega \) of possible states, equivalent to the sample space in the statistical sense, is a poset, and \( 1_{s \preceq \omega} = 1 \) if \( s \preceq \omega \) and \( 0 \) otherwise. If we index \( \Omega \) as \( \Omega = \{ \omega_{1}, \omega_{2}, \ldots, \omega_{|\Omega|} \} \), we obtain the following matrix form:

\[
\begin{align*}
\log p &\equiv F\theta - \psi(\theta),
\end{align*}
\]

where \( p \in (0, 1)^{|\Omega|} \) with \( p_i = p(\omega_i) \), \( \theta \in \mathbb{R}^{|\Omega|} \) such that \( \theta_i = \theta_{\omega_i} \) if \( \omega_i \in S \) and \( \theta_i = 0 \) otherwise, \( F = (f_{ij}) \in \{0, 1\}^{(|\Omega|) \times |\Omega|} \) with \( f_{ij} = 1_{\omega_j \preceq \omega_i} \), and \( \psi(\theta) = (\psi(\theta), \ldots, \psi(\theta)) \in \mathbb{R}^{|\Omega|} \). Each vector is treated as a column vector, and \( \log \) is an element-wise operation. This matrix form is often used as a general form of the log-linear model [Coull and Agresti 2003] and \( F \) is called a model matrix, which represents relationships between states. The assumption of the log-linear model is that \( F \) is needed to be non-singular, and [Sugiyama et al. 2017] showed that Equation (2) with a poset \( \Omega \) always provides a non-singular model matrix; that is, \( F \) is regular as long as each entry is given as \( f_{ij} = 1_{\omega_j \preceq \omega_i} \).

By inspecting Equation (2), we can see that the log-linear model belongs to the exponential family. In particular, each \( \theta_s \) corresponds to the natural parameter in the exponential family, \( \psi(\theta) \) represents the normalization constant, and \( \omega \in \Omega \) represents the outcome of each state.

The joint distribution for BSS is described by the log-linear model in Equation (2) over the sample space \( \Omega = \{ \perp \} \cup A \cup Z \cup \mathcal{X} \) equipped with the partial order defined in Equation (1). In addition, it is always assumed that the parameter space of the log-linear model \( S = A \cup Z \cup \Omega \), meaning that mixing and source layers are used as parameters to represent distributions in our model. If we learn the joint distribution from a received signal \( X \), we obtain probabilities on the source layer \( p(z_{11}), \ldots, p(z_{NM}) \), which represents normalized source signals. The rational of our approach is given as follows: The connections between each layer is structured so that the log-linear model performs a similar computation to the ICA based approach \( X = AZ \). Our structure ensures that each \( p(x_{lm}) \) is determined by \( \{ \theta_{a_{lm}} \}_{n \in [N]} \) and \( \{ \theta_{z_{nm}} \}_{l \in [L]} \) with \( [N] = \{1, \ldots, N\} \), as we always have \( a_{lm} \preceq x_{lm} \) and \( z_{nm} \preceq x_{lm} \). Moreover, more complex interaction than affine transformation, such as higher-order interactions, between signals can be treated if we additionally include partial order structure into \( Z \) and/or \( A \). These cannot be treated by a simple matrix multiplication.

Since a poset can be also represented as a directed acyclic graph (DAG), the log-linear model on a poset has a close relationship to that with a hypergraph [Ay et al. 2017] (Section 2.9). If we treat a poset as a DAG, each node of a DAG is a state of sample space and edges represent the hierarchical relationship between the states, that is, a path from a node \( \omega \) to a node \( \omega' \) exists if and only if \( \omega \preceq \omega' \). Note that this graph structure should not be confused with the graph structure found in Markov Random Fields (MRF) (undirected graph) or Bayesian Networks (directed graph), where each node typically represents a random variable. A poset forms a simplicial complex that uses its combinatorial properties to represent the higher-order interaction effects in the model [Ay et al. 2017] Definition 2.13].

### 2.3 Optimization

We train the log-linear model by minimizing the KL divergence from an empirical distribution \( \hat{p} \), which is identical to the normalized received signal \( X \in \mathbb{R}^{L \times M} \), to the

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\(^2\)We use the same symbol for an entry \( x_{lm} \) of \( X \) and its corresponding state in \( \mathcal{X} \) to avoid complicated notations.
model distribution \(p\) given by Equation (2) or, equivalently, maximizing the likelihood. More precisely, we normalize a given \(X\) by dividing each entry by the sum of all entries; that is, an empirical distribution \(\hat{p}\) is obtained as \(\hat{p}(x_{lm}) = \frac{x_{lm}}{\sum_{l,m} x_{lm}}\). If \(X\) contains negative values, an exponential kernel \(\exp(x_{lm})/\sum_{l,m} \exp(x_{lm})\) or min-max normalization \((x_{lm} + \epsilon - \min(X))/(\max(X) + \epsilon - \min(X))\) can be used, where \(\epsilon\) is some arbitrary small value to avoid zero probability. We also assumed that \(\hat{p}(a_{lm}) = 0\) and \(\hat{p}(z_{nm}) = 0\) for all \(a_{lm} \in A\) and \(z_{nm} \in Z\). These transformations do not have a negative effect on the result of the model, because we then apply the reverse transformation on the reconstructed signal or the source signal.

The objective function is given as

\[
\arg\min_{p \in \Psi_\theta} D_{\text{KL}}(\hat{p} \| p) = \arg\min_{p \in \Psi_\theta} \sum_{\omega \in \Omega} \hat{p}(\omega) \log \frac{\hat{p}(\omega)}{p(\omega)},
\]

where \(\Psi_\theta\) is the set of distributions that can be represented by Equation (2) with our structured sample space \(\Omega = \{\perp\} \cup A \cup Z \cup X\) and \(S = A \cup Z\).

The remarkable property of our model is that this optimization problem is convex and it is guaranteed that gradient-based methods can always arrive at the globally optimal solution. To show this, we analyze the geometric structure of the statistical manifold, the set of probability distributions, generated by the log-linear model. Let \(\Omega^+ = \Omega \setminus \{\perp\}\). First we introduce another parameterization \((\eta_\omega)_{\omega \in \Omega^+}\) of the log-linear model, which is defined as

\[
(\eta_\omega)_{\omega \in \Omega^+} = \sum_{s \in \Omega} 1_{\omega \succeq s} p(s).
\]

Note that \(\eta_\perp = 1\) always holds and we do not include it as a parameter. In addition, for theoretical consistency we change the parameter space used in Equation (2) from \(S\) to \(\Omega^+\) and assume that \(\theta_\perp = 0\) if \(\omega \not\in S\). Again we do not include \(\theta_\perp\) as a parameter as it is the partition function. Two parameters \((\theta_\omega)_{\omega \in \Omega^+}\) and \((\eta_\omega)_{\omega \in \Omega^+}\) have clear statistical interpretation as it is widely known that any log-linear model belongs to the exponential family, where \(\theta\) and \(\eta\) correspond to natural and expectation parameters, respectively. \(\theta\) and \(\eta\) are connected via a Legendre transformation which means that they are both differentiable and have a one-to-one correspondence. To simplify the notation, we denote by \(\theta\) and \(\eta\) the corresponding \(\theta\) and \(\eta\) of the empirical distribution \(\hat{p}\). Let

\[
\Psi = \{p \mid 0 < p(\omega) < 1 \text{ for all } \omega \in \Omega\}
\]

be the set of all probability distributions.

This set forms a statistical manifold with a dually flat structure, which is the canonical geometric structure in information geometry [Amari 2016], with its dual coordinate system \((\theta_\omega)_{\omega \in \Omega^+}, (\eta_\omega)_{\omega \in \Omega^+}\); that is, both \((\theta_\omega)_{\omega \in \Omega^+}\) and \((\eta_\omega)_{\omega \in \Omega^+}\) work as coordinate systems and determine a distribution in \(\Psi\). The Riemannian metric with respect to \(\theta\) is given as

\[
|g_{\theta_i \theta_j}| = \frac{\partial \log p(\omega)}{\partial \theta_i} \frac{\partial \log p(\omega)}{\partial \theta_j} = \sum_{\omega \in \Omega} 1_{s \preceq \omega} 1_{s' \preceq \omega} p(\omega) - \eta_s \eta_{s'},
\]

which coincides with the Fisher information [Sugiyama et al. 2017, Theorem 3] and we use it for natural gradient.

Now we consider two submanifolds \(\Psi_\theta, \Psi_\eta \subseteq \Psi\), which we define as

\[
\Psi_\theta = \{p \in \Psi \mid \theta_\omega = 0, \forall \omega \in \mathcal{E}\}, \quad \mathcal{E} = \Omega^+ \setminus \mathcal{S},
\]

\[
\Psi_\eta = \{p \in \Psi \mid \eta_\omega = \eta_\omega, \forall \omega \in \mathcal{M}\}, \quad \mathcal{M} = S.
\]

Note that this \(\Psi_\theta\) coincides with that in Equation (3). The submanifold \(\Psi_\theta\) is called an e-flat submanifold and \(\Psi_\eta\) an m-flat submanifold in information geometry. The highlight of considering these two types of submanifolds is that, if \(\mathcal{E} \cap \mathcal{M} = \emptyset\) and \(\mathcal{E} \cup \mathcal{M} = \Omega^+\), it is theoretically guaranteed that the intersection \(\Psi_\theta \cap \Psi_\eta\) is always a singleton and it is the optimizer of Equation (3) [Amari 1998, Theorem 3], that is, it is the globally optimal solution of our model.

Optimization is achieved by e-projection, which seeks \(\Psi_\theta \cap \Psi_\eta\) in the e-flat submanifold \(\Psi_\theta\). The e-projection is always convex optimization as \(\Psi_\theta\) is convex with respect to \(\theta\); this is because \(\theta\) is a coordinate system of \(\Psi_\theta\) that is linearly constrained on \(\theta\). We can therefore use the standard gradient descent strategy to optimize the log-linear model. The derivative of the KL divergence with respect to \(\theta\) is known to be the difference between expectation parameters \(\eta\) [Sugiyama et al. 2017, Theorem 2]:

\[
\frac{\partial}{\partial \theta_\omega} D_{\text{KL}}(\hat{p} \| p) = \eta_\omega - \hat{\eta}_\omega = \Delta \eta_\omega,
\]

and the KL divergence \(D_{\text{KL}}(\hat{p} \| p)\) is minimized if and only if \(\eta_\omega = \hat{\eta}_\omega\) for all \(s \in S\).

From our definition of \(\Omega\) in Equation (1), we have \(\eta_\omega = \eta_{z_{kl}}\) for all \(z_{kl}, z_{k'l'} \in Z\). Therefore all elements in the source layer will learn the same value. This problem can be avoided by removing some of partial orders between source and received layers. We propose to systematically remove the partial order \(z_{ij} \preceq z_{i'j'}\) if \(i \neq i'\) to ensure \(\eta_{z_{kl}} \neq \eta_{z_{k'l'}}\) (see Figure 1), while other strategies are possible as long as \(\eta_{z_{kl}} \neq \eta_{z_{k'l'}}\) is satisfied, for example, random deletion.

Using the above results, gradient descent can be directly applied to achieve Equation (3). However, this may need a large number of iterations to reach convergence. To reduce the number of iterations, we propose to use natural gradient [Amari 1998], which is a second-order optimization approach and will also always find the global optimum. Let us re-index \(S = A \cup Z\) as \(S = \{s_1, s_2, \ldots, s_{|S|}\}\) and assume that \(\theta = [\theta_{s_1}, \ldots, \theta_{s_{|S|}}]^T\) and \(\eta = [\eta_{s_1}, \ldots, \eta_{s_{|S|}}]^T\).
Algorithm 1 Information Geometric BSS

1: Function IGBSS(X, S):
2: Compute \( \hat{p} \) from X
3: Compute \( \hat{\eta} = (\hat{\eta}_s)_{s \in S} \) from \( \hat{p} \)
4: Initialize \((\theta_s)_{s \in S}\) (randomly or \( \theta_s = 0 \))
5: repeat
6: Compute \( p \) using the current parameter \((\theta_s)_{s \in S}\)
7: Compute \((\hat{\eta}_s)_{s \in S}\) from \( p \)
8: \((\Delta \eta_s)_{s \in Z} \leftarrow (\hat{\eta}_s)_{s \in Z} - (\hat{\eta}_s)_{s \in Z}\)
9: \((\Delta \eta_s)_{s \in A} \leftarrow (\hat{\eta}_s)_{s \in A} - (\hat{\eta}_s)_{s \in A}\)
10: Compute the Fisher information matrix for source layer \( G_Z \) and the mixing layer \( G_A \)
11: \((\theta_s)_{s \in Z} \leftarrow (\theta_s)_{s \in Z} - G_Z^{-1}(\Delta \eta_s)_{s \in Z}\)
12: \((\theta_s)_{s \in A} \leftarrow (\theta_s)_{s \in A} - G_A^{-1}(\Delta \eta_s)_{s \in A}\)
13: until convergence of \((\theta_s)_{s \in S}\)
14: End Function

In each step of natural gradient, the current \( \theta \) is updated to \( \theta_{next} \) by the following formula:

\[
\theta_{next} = \theta - G^{-1}(\eta - \hat{\eta})
\]

where \( G = (g_{ij}) \in \mathbb{R}^{(|S| \times |S|)} \) is the Fisher information matrix such that each \( g_{ij} \) is given as \( g_{s_i s_j} \) in Equation (6).

Although the natural gradient requires less iterations than the gradient descent, matrix inversion \( G^{-1} \) is computationally expensive as it has \( O(|S|^3) \) complexity. In addition, FIM values are often too small and optimization becomes numerically unstable. To solve these problems, we separate the update steps in the source and the mixing layers:

\[
(\theta_{\omega, next})_{s \in Z} = (\theta_{\omega})_{s \in Z} - G_Z^{-1}(\Delta \eta_s)_{s \in Z},
\]

\[
(\theta_{\omega, next})_{s \in A} = (\theta_{\omega})_{s \in A} - G_A^{-1}(\Delta \eta_s)_{s \in A},
\]

where \( G_Z \) and \( G_A \) are the Fisher information matrices for source and mixing layers, respectively. Note that this also leads to the same global optimum. They are constructed by assuming all the other parameters are fixed. This approach reduces the time complexity to \( O(|S|^3 + |A|^3) \).

2.4 PARAMETER COMPUTATION FOR EACH LAYER

In the following, we give \( p, \eta \), and the gradient for each layer, which are used in gradient descent.

**Received Layer (Input Layer):** Probability \( p(x) \) on the received layer \( x \in X \) is obtained as

\[
\log p(x) = \sum_{z \in Z} 1_{z \leq x} \theta_z + \sum_{a \in A} 1_{a \leq x} \theta_a + \theta_\perp,
\]

(10)

\[
\eta_x = \sum_{x' \in X} 1_{x \leq x'} p(x') = p(x).
\]

(11)

We do not need to compute gradient for this layer as there is no parameter on this layer and \( \theta_x = 0 \) for all \( x \in X \).

**Source Layer (Output Layer):** Probability \( p(z) \) on the source layer for each \( z \in Z \) is given as

\[
\log p(z) = \sum_{x' \in X} 1_{x' \leq z} \theta_x + \sum_{a \in A} 1_{a \leq z} \theta_a + \theta_\perp,
\]

(12)

\[
\eta_z = \sum_{x \in X} 1_{x \leq z} p(x) + \sum_{z' \in Z} 1_{z' \leq z} p(z')
\]

\[
= \sum_{x \in X} 1_{x \leq z} p(x) + p(z).
\]

Thus the gradient for the source layer is given as

\[
\frac{\partial}{\partial \theta_z} D_{KL}(\hat{p} || p) = \eta_z - \hat{\eta}_z
\]

\[
= \sum_{x \in X} 1_{x \leq z} (p(x) - \hat{p}(x)) + p(z).
\]

**Mixing Layer:** Probability \( p(a) \) on this layer is given as

\[
\log p(a) = \sum_{a' \in A} 1_{a \leq a'} \theta_a + \theta_\perp = \theta_a + \theta_\perp,
\]

(13)

\[
\eta_a = \sum_{x \in X} 1_{x \leq z} p(x) + \sum_{z \in Z} 1_{a \leq z} p(z)
\]

\[
+ \sum_{a' \in A} 1_{a \leq a'} p(a')
\]

\[
= \sum_{x \in X} 1_{x \leq z} p(x) + \sum_{z \in Z} 1_{a \leq z} p(z) + p(a).
\]

(14)

The gradient of the mixing layer is given as

\[
\frac{\partial}{\partial \theta_a} D_{KL}(\hat{p} || p) = \eta_a - \hat{\eta}_a
\]

\[
= \sum_{x \in X} 1_{x \leq z} (p(x) - \hat{p}(x)) + \sum_{z \in Z} 1_{a \leq z} p(z) + p(a).
\]

(15)

Parameter values \( \theta_a \) in the mixing layer represent the degree of mixing between source signals. Hence they can be used to perform feature selection and extraction. For example, if \( \theta_a = 0 \) in the extreme case, the corresponding node \( a \) does not have any contribution to the source mixing.
3 EXPERIMENTS

We empirically examine the effectiveness of IGBSS to perform BSS using real-world image and synthetic time-series datasets for an affine transformation and higher-order interactions between signals. All experiments were run on CentOS Linux 7 with Intel Xeon CPU E5-2623 v4 and Nvidia QuadroGP100.

3.1 BLIND SOURCE SEPARATION FOR AFFINE TRANSFORMATIONS ON IMAGES

In our experiments, we use three benchmark images widely used in computer vision from the University of Southern California’s Signal and Image Processing Institute (USC-SIPI) which include “airplane (F-16)”, “lake” and “peppers”. Each image is standardized to have 32x32 pixels with red, green and blue color channels with integer values between 0 and 255 to represent the intensity of each pixel. These images shown in Figure 2 are the source signal $Z$ which are unknown to the model. They are only used as ground truth to evaluate the model’s output. The equation $X = AZ$ is used to generate the received signal $X$ by randomly generating values for a mixing matrix $A$ using the uniform distribution which generates real numbers between 1 and 6. The images are then rescaled to integer values within the range between 0 and 255. The received signal $X$, which is the input to the model, is the three images shown in Figure 2b. The three images for the mixed signal may look visually similar, however, they are actually superposition of the source signal with different intensity. The objective of our model is to reconstruct the source signal $Z$ without knowing $A$.

We compare our approach to FastICA [Hyvärinen and Oja, 2000] with the log cosh function as the signal prior, dictionary learning (DL) [Olshausen and Field, 1997] with constraint for positive dictionary and positive code, and NMF with the coordinate descent solver and non-negative double singular value decomposition (NNDSVD) initialization [Boutsidis and Gallopoulos, 2008] with zero values replaced with the mean of the input.

Since BSS is an unsupervised learning problem, the order of the signal is not recovered. We identify the corresponding signal by taking all permutations of the output and calculate the minimum Euclidean distance with the ground truth. The permutation which returns the minimum error is considered as the correct order of the image. The scale of the output is also not recovered, thereby we have used min-max normalization to the output of each model.

Separation results for images are shown in Figure 2. Our method IGBSS can recover majority of the “shape” of the source signal, while the intensity of each image appears to be larger than the ground truth for all images. Small residuals of each image can be seen on the other images. For instance, in the airplane (F-16) image, residuals from the lake image can be clearly seen. Compared to the reconstruction of IGBSS with FastICA, DL and NMF, IGBSS performs significantly better as all the other approaches are unable to clearly separate the mixed signal. FastICA was unable to provide a reasonable reconstruction with 3 mixed signal. To overcome this limitation of FastICA, we randomly generated another column of the mixing matrix and append it to the current mixing matrix to create 4 mixed signals as an input to FastICA to recover a more reasonable signal.

The root mean square error (RMSE) of the Euclidean distance and the signal-to-noise ratio (SNR) between the reconstruction and the ground truth is calculated to quantify results of each method. The SNR is computed by $\text{SNR}_{dB} = 20 \log_{10}(z_{\text{norm}}/|z - z_{\text{norm}}|)$. The full results are shown in Table 1 (top row for each experiment). In the table, we present three experiments with different RGB images from USC-SIPI dataset, for each experiment we generate a new mixing matrix, where the second and the third experiments uses images of “mandrill”, “splash”, “jelly beans” and “mandrill”, “lake”, “peppers”, respectively. Our results clearly show that IGBSS is superior to other methods, that is, IGBSS has consistently produced the lowest RMSE error for every experiment. When looking at the SNR ratio, our model has produced the highest SNR for the majority of the cases and is always able to recover the same result after each run as it is formulated as a convex optimization.

3.2 BLIND SOURCE SEPARATION WITH HIGHER-ORDER FEATURE INTERACTIONS

In any real-world application, the interaction between signals are usually more complex than an affine transformation. We demonstrate the ability of BSS for our model to include higher-order feature interactions in BSS. We use the same benchmark images in the standard BSS as the source signal $Z$ for our experiment. We generate the higher-order feature interactions of the received signal by using the multiplicative product of the source signal. If we take into account up to $k$th order interaction ($k \leq N$),

$$x_{lm} = \sum_n a_{ln} z_{nm} + \sum_{n_1} \sum_{n_2 > n_1} a_{ln_1 n_2} z_{n_1 m} z_{n_2 m} + \sum_{n_1} \sum_{n_2 > n_1} \sum_{n_3 > n_2} a_{ln_1 n_2 n_3} z_{n_1 m} z_{n_2 m} z_{n_3 m} + \cdots + \sum_{n_1} \cdots \sum_{n_k > n_{k-1}} a_{ln_1 \ldots n_k} z_{n_1 m} \cdots z_{n_k m}.$$

All the other known approaches take into account only first order interactions (that is, affine transformation) between

[1] https://github.com/sjmluo/IGLLM
features. Differently, our model can directly incorporate the higher-order features as we do not assume that they are an affine transformation. When we consider up to $k$th order interactions, we additionally include the elements corresponding to new mixing parameters into the mixing layer. For example, if $k = 2$, nodes for $a_{|n_1,n_2|}$ are added and $a_{|n_1,n_2|} \leq z_{nm}$ if $n_1 = n$ or $n_2 = n$. Figure 2 shows experimental results for the third-order feature experiment. Our approach IGBSS shows superior reconstruction of the source signal to other approaches. All the other approaches except for NMF is able to achieve reasonable reconstruction. NMF is able to recover the “shape” of the image, however, unlike IBSS, NMF is a degenerate approach, so it is unable to recover all color channels in the correct proportion, creating discoloring for the image which is clearly shown in the SNR values. Since the proportion of the intensity of the pixel is not recovered. In terms of both of the RMSE and the SNR shown in Table 1 IGBSS again shows the best results for both second- and third-order interactions of signals across the three experiments.

### 3.3 TIME SERIES DATA ANALYSIS

We demonstrate the effectiveness of our model on time series data. In our experiments, we create three signals with 500 observations each using the sinusoidal function, sign function, and the sawtooth function. The synthetic data simulates typical signals from a wide range of applications including audio, medical and sensors. We randomly generate a mixing matrix by drawing from a uniform distribution with values between 0.5 and 2. In our experiment, we provide comparison of using both min-max normalization and exponential kernel as a pre-processing step and compare our approach with FastICA.

Experimental results are illustrated in Figure 3. These results show that IGBSS is superior to all the ICA approaches because it is able to recover both the shape of the signal and the sign of the signal, while all the other ICA approaches are only able to recover the shape of the signal and are unable to recover the sign of the signal. This means that ICA could recover a flipped signal. We have paired the recovered

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Table 1: Signal-to-Noise Ratio of reconstructed signal. Results for both second- and third-order interactions of signals across the three experiments.

<table>
<thead>
<tr>
<th>Exp. Order</th>
<th>Signal-to-noise (SNR) (units in dB)</th>
</tr>
</thead>
<tbody>
<tr>
<td></td>
<td>IGBSS</td>
</tr>
<tr>
<td>First*</td>
<td>12.588 ± 0.000</td>
</tr>
<tr>
<td>1</td>
<td>10.729 ± 0.000</td>
</tr>
<tr>
<td>2</td>
<td>10.612 ± 0.000</td>
</tr>
<tr>
<td>3</td>
<td>10.171 ± 0.000</td>
</tr>
</tbody>
</table>

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Table 2: Quantitative results for time-series separation experiment (mean ± standard deviation with 40 runs).

(a) Root Mean Squared Error (RMSE)

(b) Signal-to-noise (SNR) (units in dB)

<table>
<thead>
<tr>
<th>Order</th>
<th>IGBSS (min-max)</th>
<th>IGBSS (exp)</th>
<th>FastICA</th>
<th>IGBSS (min-max)</th>
<th>IGBSS (exp)</th>
<th>FastICA</th>
</tr>
</thead>
<tbody>
<tr>
<td>First</td>
<td>0.702 ± 0.000</td>
<td>0.703 ± 0.000</td>
<td>0.414 ± 0.286</td>
<td>3.596 ± 0.000</td>
<td>3.600 ± 0.000</td>
<td>15.391 ± 3.813</td>
</tr>
<tr>
<td>Second</td>
<td>0.921 ± 0.000</td>
<td>0.921 ± 0.000</td>
<td>1.700 ± 0.167</td>
<td>0.291 ± 0.000</td>
<td>0.042 ± 0.000</td>
<td>-5.803 ± 1.124</td>
</tr>
<tr>
<td>Third</td>
<td>0.967 ± 0.000</td>
<td>0.961 ± 0.000</td>
<td>1.388 ± 0.178</td>
<td>0.340 ± 0.000</td>
<td>0.128 ± 0.000</td>
<td>-3.427 ± 1.249</td>
</tr>
</tbody>
</table>
signal of ICA with the ground truth by finding the signal and sign with the lowest RMSE error. In any practical application, this is not possible for ICA because the latent signal is unknown. Through visual inspection, IGBSS is able to recover all visual signals with high accuracy, while FastICA is only able to recover the first-order interaction and it is unable to produce a reasonable recovery for second- and third-order interactions. In addition to our visual comparison, we have also performed a quantitative analysis on the experimental results using RMSE error with the ground truth. Results are shown in Table 2. FastICA has shown to have better performance for First-Order interactions. However, for second- and third-order SNR results for FastICA is unable to recover a reasonable signal because the noise is more dominant. IGBSS has shown superior performance and is able to recover the signal for second- and third-order interactions with better scores for both RMSE and SNR.

### 3.4 Runtime Analysis

In our experiment, we used a learning rate of 1.0 for gradient descent. Although the time complexity for each iteration of natural gradient is $O(|Z|^3 + |A|^3 + |\Omega||S|)$, which is larger than $O(|\Omega||S|^2)$ for gradient descent, natural gradient is able to reach convergence faster because it has quadratic convergence and requires significantly less iterations compared to gradient descent, which linearly converges. Increasing the size of the input will increase the size of $|\Omega|$ only, while the number of parameters $|Z|, |A|$ remain the same. Since the complexity of natural gradient is linear with respect to the size $|\Omega|$ of the input, increasing $|\Omega|$ does not increase the runtime significantly. Our experimental analysis in Figure 5 supports this analysis: our model scales linearly for both natural gradient and gradient descent when increasing the order of interactions in our model. This is because for practical application it is unlikely that $|A| > |Z|$. The runtime difference between natural gradient and gradient descent becomes larger as the order of interactions increases.

Finally, we compare the time complexity of our approach to the baseline approaches. NMF is typically NP-hard with an exponential runtime complexity of $O(2^NLM)$ with respect to $N$ per iteration. Note that $N$ is usually small in BSS. Similarly, the time complexity of DL using K-SVD is $O(L^2M)$ per iteration. Both the complexity of NMF and DL depends on $M$, which may be large for some applications of BSS. FastICA is usually considered to be the fastest algorithm for BSS as its complexity does not depend on $M$. FastICA overcomes the issue by taking the expectation with respect to the samples before learning the mixing matrix to reduce its complexity to $O(NL)$ per iteration. In contrast, the time complexity of our approach is cubic with respect to $M$.

### 4 Conclusion

We have proposed a blind source separation (BSS) method, called *Information Geometric Blind Source Separation* (IGBSS). We have formulated our approach using the log-linear model, which enables us to introduce a hierarchical structure into its sample space to achieve BSS. We have theoretically shown that IGBSS has desirable properties for BSS such as unique recover of source signals as it solves the convex op-
timization problem by minimizing the KL divergence from mixed signals to source signals. We have experimentally shown that IGBSS recovers images and signals closer to the ground truth than ICA, dictionary learning, and NMF. Thanks to the flexibility of the hierarchical structure, IGBSS is able to separate signals with complex interactions such as higher-order interactions. Our model is superior to the other approaches because it is non-degenerate and is able to recover the sign of the signal. Since our approach is flexible and requires less assumptions than alternative approaches, it can be applied to various real world applications such as medical imaging, signal processing, and image processing.

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References


