

Low-Rank Plus Sparse Reconstruction Using Dictionary Learning For 3D-MRI

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Abstract—This work proposes a low-rank plus sparse model using dictionary learning for 3D-MRI reconstruction from down-sampling k-space data. The scheme decomposes the dynamic image signal into two parts: low-rank part L and sparse part S and then, constructing it as a constrained optimization problem. In the optimization process, a nonconvex penalty function is used to optimize the low rank part L. The sparse part S is expressed by a over-complete dictionary using blind compressed sensing and we formulate the sparsity of coefficient matrix using l_1 norm. To avoid the ill-posed of the problem, the Frobenius norm is used in dictionary. We adopt an alternate optimization algorithm to solve the problem, which cycles through the minimization of five subproblems. Finally, we prove the effectiveness of proposed method in two cardiac cine data sets. Experimental results were compared with existing L+S, L&S and BCS schemes, which demonstrate that the proposed method behaves better in removal of artifacts and maintaining the image details.

Keywords—Blind compressed sensing(BCS); dynamic magnetic resonance imaging(dMRI); low-rank plus sparse; dictionary learning; undersampled reconstruction

I. INTRODUCTION

Dynamic magnetic resonance imaging (dMRI) is mainly used to obtain the dynamic changes of organs and it is an effective assistance in medical diagnosis such as cardiac, liver and lungs imaging. One of challenging tasks of dMRI is to accomplish High spatial and temporal resolution imaging. It requires to reconstruct the images from undersampled k-space data due to the slow data acquisition of hardwares. Since the undersampled data do not meet the requirements of Nyquist sampling rate, the recovery problem is ill-posed.

The compressed sensing (CS) approaches have been applied to dMRI to address this issue recently. Compressed sensing points out that one can achieve accurate recovery from undersampled data if the signal is compressible or sparse in a certain transform domain and the sampling basis is incoherent [1][2]. Related methods like k-t FOCUSS [3] impose sparsity constraints in the Fourier transform domain. In k-t SLR [5], the author imposes the sparsity constraint in the Karhunen Louve transform domain.

In above approaches, a sparse transform domain has to be found as a prerequisite. The commonly used sparse transform domain includes discrete cosine transform, fast Fourier transform, discrete wavelet transform, Curvelet transform, etc. However, these fixed sparsifying basis are not flexible enough to deal with diverse data adaptively. To address this issue, approaches to learn the sparsifying basis from data itself were

proposed [6]-[8], which provide more flexible representation. The blind compressed sensing [9] technique combines the advantages of both compressed sensing and dictionary learning, which do not require heuristic knowledge about sparsity basis for the signal recovery and learn a dictionary adaptively.

Inspired by the L+S reconstruction model proposed by Ricardo et al. [4] and Blind Compressed Sensing(BCS) framework proposed by Sivan et al.[9], we propose a low-rank plus sparse scheme using dictionary learning for accelerated dMRI. The L+S decomposition represents the signal more efficiently in which the L part represents static background and S part represents dynamic foreground. Instead of using a nuclear norm constraint on L part, we use a generalized Huber function [10] as the rank penalty. Results shows that the nonconvex penalty outperforms the convex nuclear norm. For the S part, we replace the fixed basis with a dictionary estimated from the undersampled data using BCS scheme. Unlike usual dictionary learning algorithm, BCS framework doesn't require a large number of training data and learns the sparsifying dictionary during signal reconstruction. Then, we reinforce the sparsity constraint in S part with the dictionary, which results in less noises and more clear details in the reconstruction images.

II. METHODS

Let $M \in C^{p \times q}$ denotes the unknown image matrix to be recovered from undersampled k-space data. Here, p represents the number of elements in a single frame, q represents the number of images in the data set. The L+S model points out that the matrix M is a decomposition of low-rank part L and sparse part S. The decomposition is unique as long as the condition satisfies L part and S part is incoherent. In general, the S part is not sparse enough itself and a sparsifying basis has to be found such that S has a sparse representation on it. As a fixed sparsifying basis is not flexible enough, a better idea is to learn the basis from the data itself adaptively, which is known as dictionary learning.

A. L+S model using dictionary learning

In our proposed method, the dictionary learning method called BCS is employed to estimate the sparsifying dictionary of S component. In general, the dictionary learning process has two phases: train for the dictionary and then, the learned dictionary is employed for reconstruction. The BCS technique

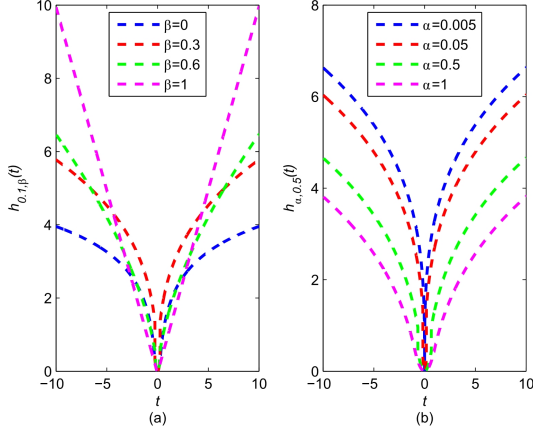


Fig. 1: (a) Huber function with α to be fixed as 0.1 (b) Huber function with β to be fixed as 0.5

combines the two phases - it estimates the sparse matrix and dictionary simultaneously during the reconstruction.

The sparse part S of the unknown images is represented as the product of two matrices - sparse matrix $\Phi_{p \times N}$ and dictionary matrix $\Psi_{N \times q}$:

$$\begin{bmatrix} \phi_1(x_1) & \dots & \phi_N(x_1) \\ \dots & \dots & \dots \\ \phi_1(x_p) & \dots & \phi_N(x_p) \end{bmatrix} \begin{bmatrix} \psi_1(t_1) & \dots & \psi_1(t_q) \\ \dots & \dots & \dots \\ \psi_N(t_1) & \dots & \psi_N(t_q) \end{bmatrix} \quad (1)$$

Here, N is the number of temporal basis in the dictionary, ϕ_i is the i th spatial coefficient vector and ψ_i is the i th temporal basis vector. N is large enough to ensure the dictionary is overcomplete, which is not necessarily orthogonal. The matrix Φ is assumed to be a sparse matrix, which means it has very few nonzero elements. Let K denotes the data acquired in k-t space and E denotes the inverse Fourier transform. Specifically, for radial or spiral k-space data, E denotes the nonuniform Fourier transform [11]. The recovery of the signal matrix M is performed by tackling the constrained optimization problem as follow:

$$\min_{L, \Phi, \Psi} \|E(L + \Phi\Psi) - K\|_2^2 + \lambda_1 \text{rank}(L) + \lambda_2 \|\Phi\|_1 \quad \text{subject to } \|\Psi\|_2 \leq \delta \quad (2)$$

where the first term denotes the data fidelity term to ensure the consistency between k-space measurement and reconstructed images, and λ_1, λ_2 are trade off parameters. The l_1 norm constraint on Φ is an approximate to l_0 norm, which ensures the sparsity of matrix Φ . The Frobenius norm constraint on Ψ is used to avoid scale ambiguity. Otherwise, the optimization will be ill-posed for the reason that matrix Φ can be arbitrarily small in magnitude.

B. The optimization Algorithm

To address the constrained optimization problem in (2), we start by introducing two proxy variables for the variables Γ, Θ respectively. Then, the rank penalty term and norm term were rewritten as follow [13]:

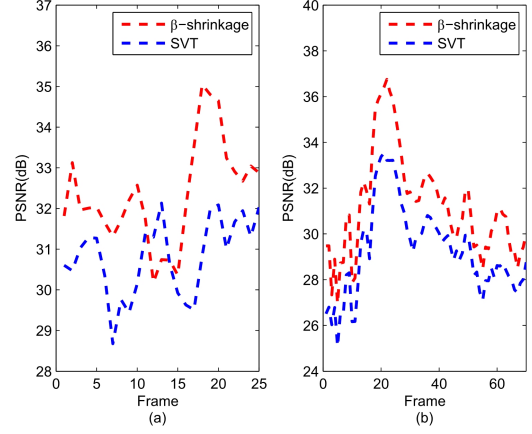


Fig. 2: Comparison of β -shrinkage and SVT of (a) data set 1 (b) data set 2

$$\min_{\Gamma} \rho_1 \|L - \Gamma\|_F^2 + \text{rank}(\Gamma) \quad (3)$$

$$\min_{\Theta} \rho_2 \|\Phi - \Theta\|_F^2 + \|\Theta\|_1 \quad (4)$$

where ρ_1, ρ_2 is a huge positive constant. Substituting (3)(4) in (2), and rewriting it as using regularization rather than strict constraints, we obtain the following modified Lagrange function :

$$\min_{L, \Gamma, \Phi, \Theta, \Psi} \|E(L + \Phi\Psi) - K\|_2^2 + \lambda_1(\rho_1 \|L - \Gamma\|_F^2 + \text{rank}(\Gamma)) + \lambda_2(\rho_2 \|\Phi - \Theta\|_F^2 + \|\Theta\|_1) + \gamma(\|\Psi\|_2 - \delta) \quad (5)$$

We solve for the object function (5) by an alternating minimization algorithm. The problem is divided into five subproblems. Each subproblem is solved by supposing all the other variables are not change. We employ an iterative approach through all subproblems until all variables convergent. For the proxy variables Γ, Θ , respective subproblems are shown in (3)(4). For the other variables, we express these subproblems as follow:

$$\min_L \|E(L + \Phi\Psi) - K\|_2^2 + \lambda_1 \rho_1 \|L - \Gamma\|_F^2 \quad (6)$$

$$\min_{\Phi} \|E(L + \Phi\Psi) - K\|_2^2 + \lambda_2 \rho_2 \|\Phi - \Theta\|_F^2 \quad (7)$$

$$\min_{\Psi} \|E(L + \Phi\Psi) - K\|_2^2 + \gamma(\|\Psi\|_2 - \delta) \quad (8)$$

Since the subproblems (6), (7), (8) are simple least squares problems, we use conjugate gradient algorithm to solve them efficiently. Subproblem (3) is an rank minimization problem. The rank operator is not convex for the rank penalty term, which implies that we need a way to relaxation. A popularly used relaxation scheme is a convex nuclear norm [14] penalty. Inspired by approach of [10], We use a nonconvex penalty using generalize Huber function instead, which outperforms a convex penalty:

$$h_{\alpha, \beta}(t) = \begin{cases} |t|^2 / 2\alpha, & \text{if } |t| < \alpha^{1/(2-\beta)} \\ |t|^\beta / \beta - \tau, & \text{if } |t| \geq \alpha^{1/(2-\beta)} \end{cases} \quad (9)$$

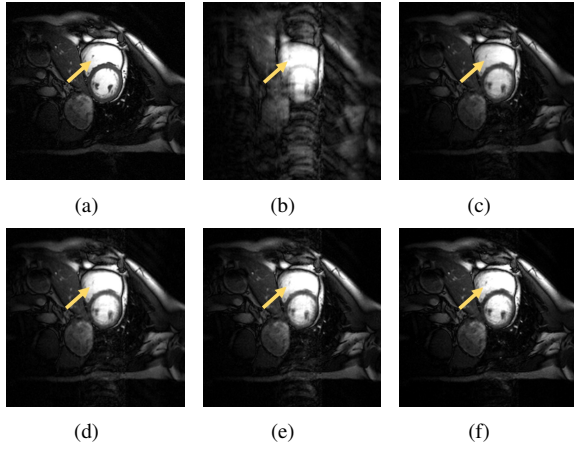


Fig. 3: data set 1 (a) ground truth (b) direct IFFT (c) L+S (d) L&S (e) BCS (f) proposed method

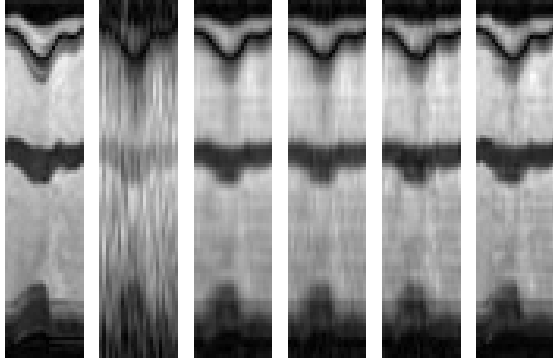


Fig. 4: Profiles of data set 1, from left to right, ground truth, direct IFFT, L+S L&S, BCS, proposed method

with $\tau = (1/\beta - 1/2)\alpha^{\beta/(2-\beta)}$. (Notice that when $\beta = 0$, $|t|^\beta/t$ becomes $\log|t|$, with $\tau = (\log\alpha - 1)/2$.) Examples of the β -Huber function is shown in Fig.1. It shows that the function is non-convex except for the region approaches the origin as $0 \leq \beta < 1$. As α increases, the convex region enlarges. Then, we define the rank penalty as following:

$$\|\Gamma\|_{h_{\alpha,\beta}} = \sum_{k=1}^{\text{Rank}(\Gamma)} h_{\alpha,\beta}(\sigma_k(\Gamma)), 0 \leq \beta \leq 1 \quad (10)$$

Note that for the case $\beta = 1$, the rank penalty becomes the popular nuclear norm. Rick [10] shows that the solution of the above optimization problem is:

$$\Gamma_{n+1} = U \text{shrink}_\beta(\Sigma, \alpha) V^H \quad (11)$$

Here, $U \Sigma V^H$ represents the SVD (singular value decomposition) of Γ_n , $\alpha = 2/\rho_1$. The β -shrinkage operation $\text{shrink}_\beta(\Sigma, \alpha)$ represents the shrinkage operator[10]:

$$\text{shrink}_\beta(t, \alpha) = \begin{cases} 0, & \text{if } |t| < \alpha |t|^{(\beta-1)} \\ (|t| - \alpha |t|^{(\beta-1)})t/|t|, & \text{if } |t| \geq \alpha |t|^{(\beta-1)} \end{cases} \quad (12)$$

Subproblem (4) is a typical l_1 norm proximal mapping problem and can be solved analytically, we solve it by performing

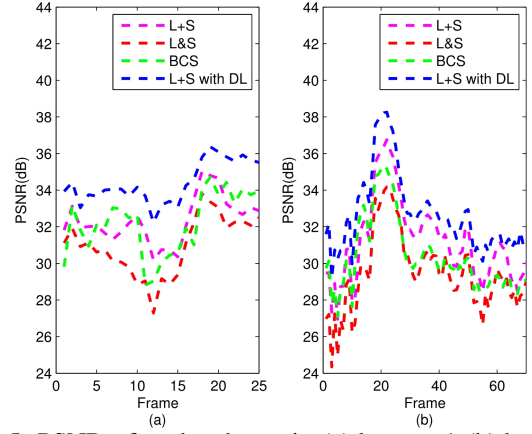


Fig. 5: PSNR of each scheme in (a) data set 1 (b) data set 2

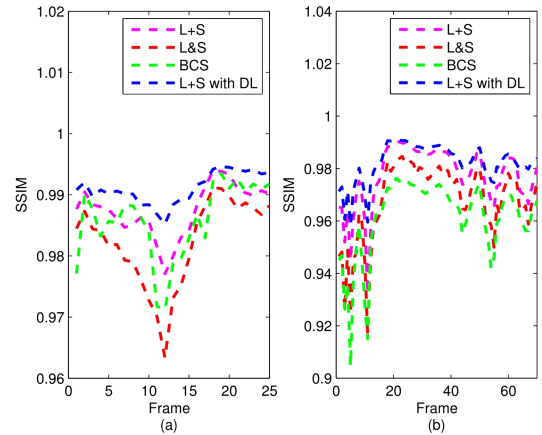


Fig. 6: SSIM of each scheme in (a) data set 1 (b) data set 2

the soft-thresholding in iterations:

$$\Theta_{n+1} = \Phi_n / |\Phi_n| / \max\{|\Phi_n| - 2/\rho_2, 0\} \quad (13)$$

We adopt a steepest descent algorithm to update the variable γ :

$$\gamma_{n+1} = \max\{\gamma_n, \gamma_n + \|\Psi\|_F^2 - \delta\} \quad (14)$$

III. EXPERIMENTAL RESULTS

A. The scheme of experiments

We used two cardiac cine data sets to evaluate our proposed method. The first Cartesian data set from a 1.5T Philips scanner contains 25 frames of ground truth [3]. The size of each frame is 256×256 , which implies 256 voxels along phase encoding direction and 256 voxels along frequency coding in the scanning. The second radial data set obtained from a Siemens 3T MRI by radial perfusion imaging [12]. For each frame, there are 256 samples per ray and 72 radial rays uniformly distribute around 360 degrees. The matrix size in image domain is $90 \times 190 \times 70$. In our experiment, the undersampling factor is 6 for the first data set and 7.5 for the second data set.

We compared the performance of proposed algorithm against three compressed sensing approaches: L+S reconstruc-

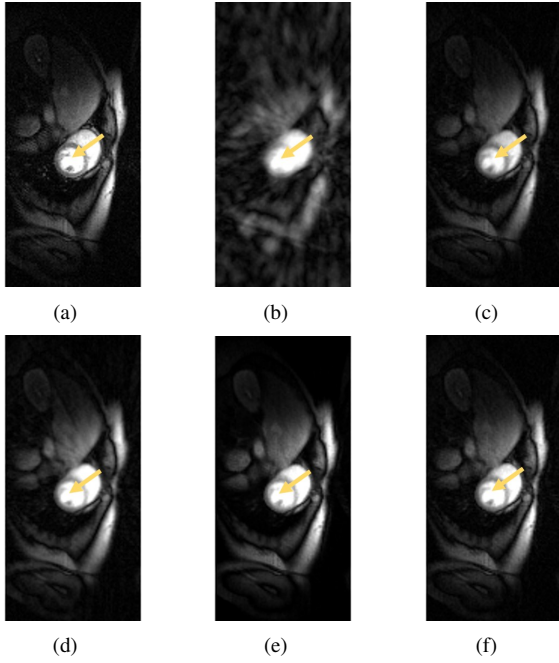


Fig. 7: data set 2 (a) ground truth (b) direct IFFT (c) L+S (d) L&S (e) BCS (f) proposed method

tion, L&S(low rank and sparse) reconstruction [15] and BCS reconstruction [12]. The implementation for L+S and BCS reconstruction were illustrated in section II and available from the authors website. The L&S approach is implemented by assuming the unknown image matrix M is low-rank and sparse simultaneously. We implemented the algorithm by ourselves.

B. Selection of parameters

The behavior of our proposed method is conditional on the selection of five parameters: smooth parameters λ_1 and λ_2 , the parameters of β -shrinkage operator α, β , the number of temporal basis N in the dictionary matrix. The selection of δ is not very important as varying it results change of regularization parameters and we set δ as 500. The parameters ρ_1, ρ_2 should be large enough as they approach to $+\infty$ theoretically and we set them as 400, which results $\alpha = 2/\rho_1 = 0.05$.

As both data sets have ground truth data, we choose the parameters λ_1, λ_2 and β such that errors between the reconstruction and ground truth is minimized. To tune the optimal parameter λ_1 , we set the λ_2 to zero and varies λ_1 in log scale. To tune the optimal parameter λ_2 , we fix λ_1 and find it by the same way. For the parameter β , we consider the case of $\beta \in \{1, 0.95, 0.9, \dots, 0\}$.

The choice of dictionary size N is refer to BCS scheme [12]. We tune the parameter N ranging from 10 to 100, and plot the reconstruction errors and nonzero coefficients in sparse matrix Φ as a function of the N . BCS scheme indicates that the reconstruction is insensitive to N as long as it is greater than a reasonable size, which we can observe from the plot.

For the data set 1, the optimal parameters are $\lambda_1 = 0.06, \lambda_2 = 0.005, \beta = 0.55, N = 65$. For the data set 2, $\lambda_1 = 0.08, \lambda_2 = 0.02, \beta = 0.7, N = 50$.

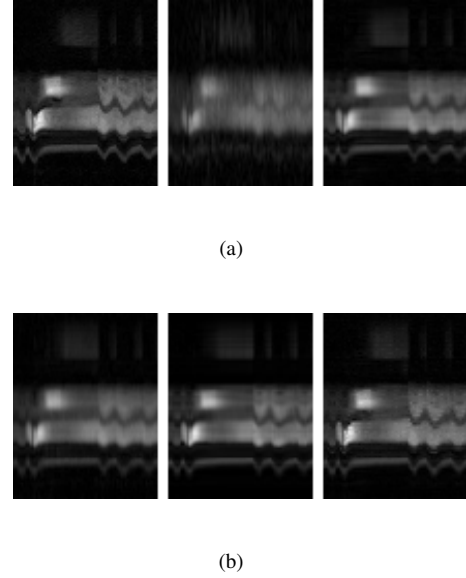


Fig. 8: From left to right, profiles of data set 2 (a) ground truth, direct IFFT, L+S (b) L&S, BCS, proposed method

C. Numerical results

We designed an experiment to validate that the non-convex rank penalty term using Huber function outperforms a convex nuclear norm penalty. We reconstructed the image sequence by L+S model [4] with β -shrinkage operation and singular-value threshold (SVT) operation respectively. The PSNR (Peak Signal-to-Noise Ratio) is popularly used in the assessment of images. Fig.2. shows that β -shrinkage operation achieves higher PSNR than SVT operation for both data sets. This proves the effectiveness of non-convex rank penalty.

Fig.3. shows the comparison of reconstruction of the fourth frame in first data set. Fig.3 (a) is ground truth as reference. The direct inverse FFT reconstruction of undersampled k-space data is badly blurred with artifacts. As arrows indicates, the L&S and BCS schemes lost some details and the L+S scheme caused shadows in local region, while The proposed method behaved better in both aspects.

data set 1	L+S	L&S	BCS	proposed
PSNR(dB)	32.3595	30.9444	32.1948	34.3995
SSIM	0.9875	0.9823	0.9858	0.9911

TABLE I: Average of NSME,PSNR,SSIM in data set 1

data set 2	L+S	L&S	BCS	proposed
PSNR(dB)	31.2019	29.4289	30.3719	32.6869
SSIM	0.9773	0.9679	0.9615	0.9823

TABLE II: Average of NSME,PSNR,SSIM in data set 2

To observe the cardiac motion in data set 1, Fig.4. shows the temporal profiles of a fixed phase encoding line along

It is difficult to observe difference of reconstructed images' quality by naked eyes. Fig.5. shows the comparison of different scheme using PSNR in both data sets. As we can see, our proposed method achieves the highest PSNR among them. This proves that with the sparse basis learned from data itself, we have a better sparsity constraint for dMRI reconstruction.

In comparison, we make quantitative analysis by calculating the average of PSNR, SSIM, NMSE and illustrates them in table I and table II. For data set 1, reconstruction errors get 0.5%-1.2% decrease and PSNR get 2.0-3.4 dB improvement and structural similarity get 0.4%-0.9% improvement by proposed method. For data set 2, reconstruction errors get 0.6%-1.8% decrease and PSNR get 1.5-3.2 dB improvement and structural similarity get 0.5%-1.0% improvement by proposed method.

IV. CONCLUSION

In the future work, the proposed method can be extended in several directions. For example, we can applicate it on 4-

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REFERENCES

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