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# An Improved Adaptive Subspace Tracking Algorithm Based on Approximated Power Iteration

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**ABSTRACT** A subspace tracking technique has drawn a lot of attentions due to its wide applications. The main objective of this approach is to estimate signal or noise subspace basis for the sample covariance matrix. In this paper, we focus on providing a fast, stable, and adaptive subspace tracking algorithm that is implemented with low computational complexity. An alternative realization of the fast approximate power iteration (FAPI) method, termed modified FAPI (MFAPI), is also presented. Rather than solving an inverse square root of a matrix employed in the FAPI, the MFAPI applies the matrix product directly to ensure the orthonormality of the subspace basis matrix at each recursion. This approach yields a simpler derivation and is numerically stable while maintaining a similar computational complexity as compared with that of the FAPI. Furthermore, we present a detailed mathematical proof of the numerical stability of our proposed algorithm. Computer simulation results indicate that the MFAPI outperforms many classical subspace tracking algorithms, particularly at the transient-state step.

**INDEX TERMS** Adaptive subspace tracking, approximated power iteration, orthonormal iteration, projection approximation.

## I. INTRODUCTION

The subspace tracking methods, that are employed to track the change of structure in signals, intend to divide the observed data into two parts: signal subspace and noise subspace. Since a more distinct representation of useful signals could be obtained in this way, these methods have drawn lots of interests recently. There are many subspace-based application in practice such as sensor networks [1], signal detection [2]–[4], machine learning [5], array processing [6], etc. In some applications, these methods can detect very weak signals in a very noisy environment at signal-to-noise ratio (SNR) as low as to -23dB [4].

The subspace tracking methods can be described as follows: let x(t) be a N-dimensional received data vector with

sample covariance matrix  $C_{xx}(t) = E[x(t)x(t)^H]$ . The main objective is then to estimate the largest L-dimensional signal subspace or the smallest (N-L)-dimensional noise subspace spanned by the basis matrix  $W(t) \in \mathbb{C}^{N \times L}$  or  $W(t) \in \mathbb{C}^{N \times (N-L)}$  of  $C_{xx}(t)$ , respectively, where L is the number of the useful signals in a noisy environment.

There exist many different methods related with subspace tracking in literature. Computational complexity measure is a commonly employed indicator to classify and distinguish these algorithms. According to [17], the computational complexity of an algorithm can be classified as high, medium, or low based on the number of operations,  $O(N^3)$ ,  $O(N^2L)$  and O(NL), respectively. Direct decomposition methods, such as batch singular value decomposition (SVD)

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and eigenvalue decomposition (EVD), are not suitable techniques in real applications due to their high computational complexity in order of  $O(N^3)$ , even though they yield quite accurate eigenvalue and eigenvector estimations. Presently, almost all mainstream subspace tracking methods have been able to reduce the computational complexity to the order O(NL) operations. In practice, these algorithms find wide range of applications [7], especially with massive data computations. Therefore, in this paper, we are mainly concerned with these mainstream low-complexity schemes.

Subspace-based algorithms are designed to track the signal (principal) or noise (minor) subspace. Generally, according to the tracking type, these methods can be categorized into two types: the algorithms for tracking only one single subspace or tracking both subspaces simultaneously. Also, most subspace tracking algorithms can be viewed as a constrained or unconstrained optimization problem [7]. Benefiting from using recursive least squares (RLS) approach, the popular types of subspace tracking methods, such as projection approximation subspace tracking (PAST) [8], orthonormal PAST (OPAST) [9], fast approximated power iteration (FAPI) [10], have very fast convergence rates in estimating the signal or noise subspace. However, these methods are not without their drawbacks. Although the method based on PAST convergences to the basis matrix, orthonormalization can not be guaranteed in the initial stage. Comparing with PAST, the OPAST has a faster convergence rate due to its orthonormalization to the signal subspace matrix at each iterative step. Both PAST and OPAST can be regarded as a first order approximation of the FAPI. Hence, FAPI outperforms PAST and OPAST. However, the main weakness of the FAPI algorithm is that its derivation is relatively difficult and its numerical instability at the transient-state step (details will be described in the following sections). For the PAST-type algorithms, a more detailed description can be found in [11]. Comparison with RLS employed in those algorithms, the FRANS algorithm can track noise subspace basis based on Rayleigh's quotient [12]. In order to overcome the instability and slow convergence rate of FRANS, a algorithm called HFRANS is proposed using Householder transformation [13] whose detailed analysis is presented in [14]. Note that, these algorithms belong to the first type.

The second type of methods, mainly inherited from the data projection method (DPM) [15], can track dual signal or noise subspace matrix simultaneously with a sign change. Popular algorithms include ODKA [16], FDPM [17], YAST [18], SGYAST [19], and algorithm proposed in [20]. Although most of the DPM-based algorithms have simple code structure and ability to track dual subspace basis matrix, they have slower convergence rate than that of the first type of algorithms.

Considering the above mentioned problems, this paper presents a new implementation of the tracking signal subspace approach based on FAPI for the exponential window case. Relying on the direct matrix product, a simple but more stable method is provided to guarantee orthonormalization of subspace basis matrix. Additionally, we discuss convergence rates of the two types of algorithms mentioned above through the results obtained by detail computer simulations. Finally, a thorough mathematical proof of the numerical stability is given which makes it easier to understand our algorithm. To the best of our knowledge, similar description has not been proposed in the literature yet.

The rest of the paper is organized as follows. In Section II, problem definition is provided; In Section III, we first review the exponential window estimate of the sample covariance matrix, then present a detailed derivation of our MFAPI algorithm; performance analyses, including computational complexity, convergence rate and numerical stability of proposed MFAPI method are presented in Section IV; finally, simulation results and conclusion are provided in Section V and Section VI, respectively.

## **II. PROBLEM DEFINITION AND NOTATION**

We first provide the notations used in this paper. Scalars are denoted by lower case Greek letters, the symbol  $\mathbb{C}^{m\times n}$  represents the complex matrix with m rows and n columns,  $I_r$  represents the  $r\times r$  identity matrix. The transpose of a matrix A is denoted by  $A^T$  and the conjugate-transpose by  $A^H$ . The symbol  $\|\cdot\|$  denotes the 2-norm of a vector,  $\|\cdot\|_F^2$  denotes the matrix Frobenius norm, diag $\{\cdot\}$  is a diagonal matrix.

In this paper, the subspace basis matrix is estimated by our proposed subspace-based algorithm, which can be applied to the various scenarios described in the introduction. However, in order to evaluate the performance of the proposed algorithm, a special example of detecting the frequencies of signals is considered.

Let x(t) be a received data vector. x(t) can be written as  $x(t) = [x_1(t), x_2(t), ..., x_N(t)]^T$  obtained from the output of N different sensors of an array at the tth snapshot, or  $x(t) = [x(t), x(t-1), ..., x(t-N+1)]^T$  obtained from the N sequentially received data of a time series. The signal model, including L complex exponential signals, can be expressed as

$$\mathbf{x}(t) = \sum_{l=1}^{L} p_l(t)\mathbf{e}(\omega_l) + \mathbf{n}(t)$$
 (1)

where  $p_l(t)$  is the random signal,  $\omega_l$  denotes the lth frequency to be detected,  $\boldsymbol{n}(t)$  is a stationary zero-mean additive white Gaussian noise (AWGN) vector with variance  $\sigma^2$ .  $\boldsymbol{e}(\omega_l) = [1 \exp(j\omega_l) \exp(j2\omega_l) \cdots \exp(j(N-1)\omega_l)]^T$  is the frequency vector. Let  $\boldsymbol{E}(\omega) = [\boldsymbol{e}(\omega_1), \boldsymbol{e}(\omega_2), ..., \boldsymbol{e}(\omega_L)]$  be the matrix of  $\omega$  and  $\boldsymbol{P} = [p_1, p_2, ..., p_L]^T$  be a vector of p. Equation (1) can then be rewritten as

$$\mathbf{x}(t) = \mathbf{E}(\omega)\mathbf{P} + \mathbf{n}(t). \tag{2}$$

Here, we assume that the signal is independent of the noise. Then, the covariance matrix of x(t) can be obtained as [15]

$$C_{xx}(t) = \mathbb{E}[\mathbf{x}(t)\mathbf{x}(t)^H] = \mathbf{E}\mathbf{P}_s\mathbf{E}^H + \sigma^2\mathbf{I}_N, \tag{3}$$



where  $P_s = E[PP^H]$  is the covariance matrix of the signal components,  $E[\cdot]$  denotes the expectation operator. By applying EVD,  $C_{xx}(t)$  can be further expressed as the sum of the signal and the noise covariance matrices as

$$C_{xx}(t) = W_s \Sigma_s W_s^H + W_{\psi} \Sigma_{\psi} W_{\psi}^H, \tag{4}$$

where  $W_s$  and  $W_{\psi}$  denote the signal and noise subspace basis matrices, respectively,  $\Sigma_s = \text{diag}\{\lambda_1, \dots, \lambda_L\}$  and  $\Sigma_{\psi} = \text{diag}\{\lambda_{L+1}, \dots, \lambda_N\}$  denote signal and noise eigenvalues, respectively.  $C_{xx}(t)$  is semi-positive definite matrix whose eigenvalues satisfy

$$\lambda_1 \ge \dots \ge \lambda_L \ge \lambda_{L+1} = \lambda_{L+2} \dots = \lambda_N = \sigma^2 \ge 0.$$
 (5)

The basis matrix  $W_{\psi}$  is employed to estimate the frequencies  $\omega_l$  of the signal components. Some algorithms such as MUSIC [21] and ESPRIT [22] can be used for this purpose. Th MUSIC algorithms is described as follows. Equation (4) can be rewritten as

$$\boldsymbol{C}_{xx}(t) = [\boldsymbol{W}_s, \boldsymbol{W}_{\psi}] \begin{bmatrix} \boldsymbol{\Sigma}_s & \mathbf{0} \\ \mathbf{0} & \sigma^2 \boldsymbol{I}_{N-L} \end{bmatrix} \begin{bmatrix} \boldsymbol{W}_s^H \\ \boldsymbol{W}_{\psi}^H \end{bmatrix}. \tag{6}$$

Considering  $W_s^H W_{\psi} = 0$  and  $W_{\psi}^H W_{\psi} = I$ , we have

$$C_{xx}(t)W_{\psi} = [W_s, W_{\psi}] \begin{bmatrix} \Sigma_s & 0 \\ 0 & \sigma^2 I_{N-L} \end{bmatrix} \begin{bmatrix} 0 \\ I \end{bmatrix} = \sigma^2 W_{\psi}. \quad (7)$$

Using (3) we derive

$$C_{xx}(t)W_{yt} = EP_{s}E^{H}W_{yt} + \sigma^{2}W_{yt}. \tag{8}$$

And from (7), (8) we have

$$\mathbf{E}\mathbf{P}_{s}\mathbf{E}^{H}\mathbf{W}_{\psi} = \mathbf{0}. \tag{9}$$

It then follows that

$$\mathbf{W}_{y_t}^H \mathbf{E} \mathbf{P}_s \mathbf{E}^H \mathbf{W}_{\psi} = \mathbf{0}. \tag{10}$$

Note that when Q is nonsingular,  $t^H Qt = 0$  if, and only if, t = 0 holds. Hence, from (10)

$$\mathbf{E}^H \mathbf{W}_{\psi} = \mathbf{0}. \tag{11}$$

Substituting  $E(\omega) = [e(\omega_1), e(\omega_2), ..., e(\omega_L)]$  into (11) yields

$$\boldsymbol{e}(\omega)^{H}\boldsymbol{W}_{\psi} = \boldsymbol{0}, \ \omega = \omega_{1}, \omega_{2}, ..., \omega_{L}.$$
 (12)

Finally, for the given  $W_{\psi}$ , let

$$p(\omega) = \frac{1}{e(\omega)^H W_{\psi} W_{\psi}^H e(\omega)}$$
(13)

be a spectral function of the parameter  $\omega$ . We can then obtain L largest peaks of  $p(\omega)$  by searching over the values of  $\omega$ . The  $\omega$ 's corresponding to the L largest peaks are the estimated frequencies.

## III. PROPOSED MFAPI ALGORITHM

FAPI has excellent capability as compared to other existing subspace-based algorithms in tracking the subspace basis matrix and, especially, in achieving better convergence rate [10]. However, the main weakness of FAPI is that it needs to evaluate the inverse square root of matrix to implement orthonormalization for subspace basis matrix. It is known that this evaluation may cause numerical instability with a substantially higher computational complexity. On the other hand, the proposed MFAPI method makes use of direct matrix multiplication for orthonormalization to obtain a faster convergence rate than the FAPI and guarantees numerical stability. In the rest of this section, a detail derivation of MFAPI algorithm based on FAPI is presented.

## A. ESTIMATE OF SAMPLE COVARIANCE MATRIX

Subspace tracking methods need to estimate their sample covariance matrix from the received data. Usually, the adapted schemes for covariance matrix estimation is based on using the exponential and truncated windowings whose details can be found in [10].

#### 1) EXPONENTIAL WINDOW ESTIMATE

In general, exponential windowing, from which  $C_{xx}(t)$  is estimated, is defined as

$$C_{xx}(t) = \sum_{k=-\infty}^{t} \beta^{t-k} \mathbf{x}(k) \mathbf{x}(k)^{H}, \qquad (14)$$

where  $\beta$  is a forgetting factor with  $0 < \beta \le 1$ . Equation (14) usually has another form for the recursive purpose and it can be written as [8]

$$C_{xx}(t) = \beta C_{xx}(t-1) + x(t)x(t)^{H}.$$
 (15)

Exponential windowing-based covariance estimation puts emphasis on the new data so it fits in non-stationary scenarios when  $\beta$  is small enough. Meanwhile, the estimate of covariance matrix using exponential windowing can be computed simply by an iterative algorithm without storing the historical sample data, that is very useful in big data applications.

# 2) TRUNCATED WINDOW ESTIMATE

Truncated windowing-based covariance estimation can be defined as [10]

$$C_{xx}(t) = \sum_{k=t-l+1}^{t} \beta^{t-k} \mathbf{x}(k) \mathbf{x}(k)^{H},$$
 (16)

where  $\beta = 1$  corresponds to the case of sliding windowing, l is the window length. For iterative purpose, truncated windowing has the following form:

$$C_{xx}(t) = \beta C_{xx}(t-1) + x(t)x(t)^{H} - \beta^{l}x(t-l)x(t-l)^{H}.$$
(17)

In this paper, for simplicity, we only consider implementation of MFAPI using an exponential windowing due to the



following reasons. First, our derivation method for MFAPI can be realized easily with the truncated windowing case. Secondly, for the FAPI, computational complexity is much lower in case of the exponential windowing than the truncated windowing. Namely, it is about 3NL (3NL is the lowest complexity among the current low-complexity subspace tracking algorithms) for the exponential windowing while 6NL for the truncated windowing.

#### **B. POWER ITERATION**

Power iteration [7] is a commonly employed method to compute dominant subspace basis matrix. Since FAPI algorithm has a similar code structure with power iteration, we describe the power iteration method first. With sequentially received data x(t), conventional power iteration method computes the subspace basis matrix W(t) by the following formulas

$$y(t) = W(t-1)^{H} x(t),$$
 (18)

$$\boldsymbol{C}_{xy}(t) = \boldsymbol{C}_{xx}(t)\boldsymbol{W}(t-1), \tag{19}$$

$$\mathbf{W}(t)\mathbf{R}(t) = \mathbf{C}_{xy}(t), \tag{20}$$

where L-dimensional y(t) can be viewed as data compressing step by projecting N-dimensional x(t) onto W(t-1) (in general,  $L \ll N$ ), and  $C_{xy}(t)$  as correlation matrix between x(t) and y(t). W(t) can be computed by the factorization step (20) by means of the QR decomposition or other orthonormalization methods. R(t) is a matrix factor of  $C_{xy}(t)$ . It is an upper triangular matrix if QR decomposition applied. Hence, the power iteration method can eventually converge to the principal subspace basis matrix.

## C. FAPI ALGORITHM

FAPI algorithm is designed for estimating subspace basis matrix with low-computational complexity and fast convergence rate. In the following, we describe the implementation of this algorithm. As discussed in [10], the following approximation is employed for projection of W(t-1) into W(t):

$$W(t) \simeq W(t-1)\Theta(t), \tag{21}$$

where  $\Theta(t) \stackrel{\triangle}{=} W(t-1)^H W(t)$  is an orthonormal matrix. Applying the exponential window estimate and substituting (15) into (19), we obtain

$$C_{xy}(t) = \beta C_{xx}(t-1)W(t-1) + x(t)y(t)^{H}.$$
 (22)

Using the approximation (21) of  $\Theta(t)$  at time (t-1), Equation (22) can be expressed as

$$\boldsymbol{C}_{xy}(t) = \beta \boldsymbol{C}_{xy}(t-1)\boldsymbol{\Theta}(t-1) + \boldsymbol{x}(t)\boldsymbol{y}(t)^{H}. \tag{23}$$

Equation (23) is one of the three important formulas that are used in [10]. Other two main iterative expressions, that are needed to estimate the subspace basis matrix W(t) in the FAPI algorithm [10], are given as

$$\mathbf{Z}(t) = \frac{1}{\beta} \mathbf{\Theta}(t)^{H} (\mathbf{I}_{L} - \mathbf{g}(t)\mathbf{y}(t)^{H}) \mathbf{Z}(t-1) \mathbf{\Theta}(t)^{-H}, \quad (24)$$

$$\mathbf{W}(t) = \left(\mathbf{W}(t-1) + \mathbf{e}(t)\mathbf{g}(t)^{H}\right)\mathbf{\Theta}(t), \tag{25}$$

where  $\mathbf{Z}(t)$  is an auxiliary matrix,  $\mathbf{e}(t)$  is projection error of  $\mathbf{x}(t)$ , and  $\mathbf{g}(t)$  is defined in [10] as

$$e(t) = x(t) - W(t-1)y(t),$$
 (26)

$$\mathbf{g}(t) = \frac{\mathbf{h}(t)}{\beta + \mathbf{v}(t)^H \mathbf{h}(t)},\tag{27}$$

where

$$\boldsymbol{h}(t) = \mathbf{Z}(t-1)\mathbf{y}(t). \tag{28}$$

Proper selection of  $\Theta(t)$  in (25) is critical for subspace tracking algorithms. The key point is to make W(t) orthonormal at each iterative step, with low computational complexity and numerically stable way. Namely

$$\boldsymbol{W}(t)^{H}\boldsymbol{W}(t) = \boldsymbol{I}_{L}. \tag{29}$$

For the FAPI algorithm,  $\Theta(t)$  is selected as an inverse square root as follows,

$$\mathbf{\Theta}(t)\mathbf{\Theta}(t)^{H} = (\mathbf{I}_{L} + \mathbf{g}(t)(\mathbf{e}(t)^{H}\mathbf{e}(t))\mathbf{g}(t)^{H})^{-1}.$$
 (30)

However, derivation of  $\Theta(t)$  is based on the assumption that  $\beta + y(t)^H h(t)$  is nonsingular and solving an inverse square root requires higher computational complexity and some instabilities are inevitable.

Based on the above mentioned discussions, we now present the derivation of the MFAPI algorithm.

## D. DERIVATION OF MFAPI

In this paper, we propose another way for selecting  $\Theta(t)$  that guarantees orthonormalization of W(t). This approach is based on using a direct matrix production. As a direct results, it provides more simpler derivation than FAPI and avoids any instability in the algorithm.

## 1) DERIVATION FOR $\theta(t)$

Let

$$T(t) \triangleq W(t-1) + e(t)g(t)^{H}. \tag{31}$$

It then follows that

$$T(t)^{H}T(t) = I_{L} + ||e(t)||^{2}g(t)g(t)^{H}.$$
 (32)

In deriving (32), we have assumed that W(t-1) is orthonormal at time (t-1), that is,  $W(t-1)^H W(t-1) = I_L$ . Additionally, it can be easily seen by simple manipulations that  $W(t-1)^H e(t) = 0_{L \times 1}$ . Let

$$Y(t) \triangleq \left[ \frac{\mathbf{g}(t)}{\|\mathbf{g}(t)\|} : \mathbf{G}(t) \right]$$
 (33)

be a  $L \times L$  column orthonormal matrix, satisfying:  $\frac{g(t)g(t)^H}{\|g(t)\|^2} + G(t)G(t)^H = I_L$ ,  $G(t)^H \frac{g(t)}{\|g(t)\|} = \mathbf{0}_{(L-1)\times 1}$  and  $G(t)^H G(t) = I_{L-1}$ . It then follows that

$$(\boldsymbol{T}(t)\boldsymbol{Y}(t))^{H}\boldsymbol{T}(t)\boldsymbol{Y}(t)$$

$$= \boldsymbol{Y}(t)^{H}\boldsymbol{T}(t)^{H}\boldsymbol{T}(t)\boldsymbol{Y}(t)$$

$$= \boldsymbol{I}_{L} + \operatorname{diag}\left\{\|\boldsymbol{e}(t)\|^{2}\|\boldsymbol{g}(t)\|^{2}, 0, \dots, 0\right\}$$



= diag 
$$\left\{ 1 + \|\boldsymbol{e}(t)\|^2 \|\boldsymbol{g}(t)\|^2, 1, \dots, 1 \right\}$$
  
= diag  $\left\{ \|\boldsymbol{v}(t)\|^2, 1, \dots, 1 \right\},$  (34)

where v(t) is defined as  $v(t) = W(t-1) \frac{g(t)}{\|g(t)\|} + e(t) \|g(t)\|$ , which satisfies

$$\|\boldsymbol{v}(t)\|^2 = 1 + \|\boldsymbol{e}(t)\|^2 \|\boldsymbol{g}(t)\|^2. \tag{35}$$

Note that the right hand side of (34) is a diagonal matrix. and can be transformed into an identity matrix by applying a normalization on the first column. A common method for orthonormalization is to multiply it by an inverse matrix, as employed in [10]. With a simple but rather more stable way, we implement normalization here by multiplying a suitable matrix having the following form

$$U(t) = \operatorname{diag}\left\{\frac{1}{\|v(t)\|}, 1, \dots, 1\right\}.$$
 (36)

For implementing orthonormalization,  $\Theta(t)$ selected according to the following relation

$$\Theta(t) = \mathbf{Y}(t)\mathbf{U}(t)\mathbf{Y}(t)^{H} 
= \left[\frac{\mathbf{g}(t)}{\|\mathbf{g}(t)\|} : \mathbf{G}(t)\right]\mathbf{U}(t) \begin{bmatrix} \frac{\mathbf{g}(t)^{H}}{\|\mathbf{g}(t)\|} \\ \dots \\ \mathbf{G}(t)^{H} \end{bmatrix} 
= \frac{\mathbf{g}(t)\mathbf{g}(t)^{H}}{\|\mathbf{v}(t)\|\|\mathbf{g}(t)\|^{2}} + \mathbf{G}(t)\mathbf{G}(t)^{H}.$$
(37)

Since  $\frac{\mathbf{g}(t)\mathbf{g}(t)^H}{\|\mathbf{g}(t)\|^2} + \mathbf{G}(t)\mathbf{G}(t)^H = \mathbf{I}_L$  holds, eliminating the term  $\mathbf{G}(t)\mathbf{G}(t)^H$  in (37) yields

$$\mathbf{\Theta}(t) = \mathbf{I}_L + (\frac{1}{\|\mathbf{v}(t)\|} - 1) \frac{\mathbf{g}(t)\mathbf{g}(t)^H}{\|\mathbf{g}(t)\|^2}.$$
 (38)

# 2) UPDATE FOR Z(t) USING $\Theta(t)$

For updating  $\mathbf{Z}(t)$  in (24), we apply the matrix inversion lemma to  $\Theta(t)$  as

$$\Theta(t)^{-1} = \left[ I_L + \left( \frac{1}{\|v(t)\|} - 1 \right) \frac{g(t)g(t)^H}{\|g(t)\|^2} \right]^{-1} 
= I_L - \frac{\left( \frac{1}{\|v(t)\|} - 1 \right) \frac{g(t)g(t)^H}{\|g(t)\|^2}}{1 + \left( \frac{1}{\|v(t)\|} - 1 \right) \frac{g(t)^H g(t)}{\|g(t)\|^2}} 
= I_L + \frac{(\|v(t)\| - 1)}{\|g(t)\|^2} g(t)g(t)^H 
= I_L + \delta(t)g(t)g(t)^H,$$
(39)

where  $\delta(t) = \frac{\|\boldsymbol{v}(t)\| - 1}{\|\boldsymbol{g}(t)\|^2}$ . Substituting (38), (39) into (24) yields

$$\mathbf{Z}(t) = \frac{1}{\beta} [\mathbf{Z}(t-1) - \mathbf{g}(t)\mathbf{h}'(t)^H + \boldsymbol{\epsilon}(t)\mathbf{g}(t)^H], \quad (40)$$

where h'(t) and  $\epsilon(t)$  are defined by

$$\mathbf{y}'(t) = \frac{1}{\|\mathbf{v}(t)\|} \left[ \delta(t)\mathbf{g}(t) + \mathbf{y}(t) \right], \tag{41}$$

$$\boldsymbol{h}'(t) = \mathbf{Z}(t-1)^H \mathbf{y}'(t), \tag{42}$$

$$\boldsymbol{\epsilon}(t) = \delta(t) \left[ \mathbf{Z}(t-1)\mathbf{g}(t) - \mathbf{g}(t)\mathbf{h}'(t)^{H}\mathbf{g}(t) \right]. \tag{43}$$

# 3) UPDATE FOR W(t) USING $\Theta(t)$

Orthonormalization of W(t) needs to be implemented exactly at each iterative step. Using  $\Theta(t)$  in (38), W(t) in (25) can be computed as follows. Substituting (38) into (25) yields

$$W(t) = W(t-1) + \frac{1}{\|\boldsymbol{v}(t)\|} \boldsymbol{e}(t) \boldsymbol{g}(t)^{H}$$

$$+ (\frac{1}{\|\boldsymbol{v}(t)\|} - 1) W(t-1) \frac{\boldsymbol{g}(t) \boldsymbol{g}(t)^{H}}{\|\boldsymbol{g}(t)\|^{2}}$$

$$= W(t-1) + \boldsymbol{e}'(t) \boldsymbol{g}(t)^{H},$$
(44)

where

$$e'(t) = \frac{1}{\|\boldsymbol{v}(t)\|} e(t) + (\frac{1}{\|\boldsymbol{v}(t)\|} - 1) \frac{\boldsymbol{W}(t-1)\boldsymbol{g}(t)}{\|\boldsymbol{g}(t)\|^2}.$$
 (45)

Substituting (41) into (45) yields

$$e'(t) = \frac{1}{\|v(t)\|} x(t) - W(t-1)y'(t).$$
 (46)

After some algebra, it can be shown that

$$W(t)^{H}W(t) = \mathbf{\Theta}(t)^{H}T(t)^{H}T(t)\mathbf{\Theta}(t)$$

$$= Y(t)U(t)^{H}Y(t)^{H}T(t)^{H}T(t)Y(t)U(t)Y(t)^{H}$$

$$= I_{L}.$$
(47)

## TABLE 1. Pseudo-code of the MFAPI algorithm.

$$m{W}(0) = egin{bmatrix} m{I}_L \ m{0}_{(N-L) imes L} \end{bmatrix}, m{Z}(0) = m{I}_L$$

Input sequential data vector: $x(t)$		Cost
$oldsymbol{y}(t) = oldsymbol{W}(t-1)^H oldsymbol{x}(t)$	(18)	$NL$ $L^2$
$\boldsymbol{h}(t) = \boldsymbol{Z}(t-1)\boldsymbol{y}(t)$	(28)	$L^2$
$oldsymbol{g}(t) = rac{oldsymbol{h}(t)}{eta + oldsymbol{y}(t)^H oldsymbol{h}(t)}$	(27)	

$$||e(t)||^2 = ||x(t)||^2 - ||y(t)||^2$$

$$* * ||v(t)|| = \sqrt{1 + ||e(t)||^2 ||g(t)||^2}$$
(26)  $N + L$ 

$$L$$

\*\* 
$$\|v(t)\| = \sqrt{1 + \|e(t)\|^2 \|g(t)\|^2}$$

$$\delta(t) = \frac{\|v(t)\|^{-1}}{\|g(t)\|^2}$$

$$y'(t) = \frac{1}{\|v(t)\|} [\delta(t)g(t) + y(t)]$$
(41)  $L$ 

$$\mathbf{y}'(t) = \frac{1}{\|\mathbf{v}(t)\|} \left[ \delta(t)\mathbf{g}(t) + \mathbf{y}(t) \right]$$
 (41)  $L$ 

$$\boldsymbol{h}'(t) = \boldsymbol{Z}(t-1)^{H} \boldsymbol{y}'(t) \tag{42}$$

$$\boldsymbol{\epsilon}(t) = \delta(t) \left[ \boldsymbol{Z}(t-1)\boldsymbol{g}(t) - \boldsymbol{g}(t)\boldsymbol{h}'(t)^{H}\boldsymbol{g}(t) \right]$$
(43) 
$$\boldsymbol{L}^{2} + 3L$$
$$\boldsymbol{Z}(t) = \frac{1}{2} \left[ \boldsymbol{Z}(t-1) - \boldsymbol{g}(t)\boldsymbol{h}'(t)^{H} + \boldsymbol{\epsilon}(t)\boldsymbol{g}(t)^{H} \right]$$
(40) 
$$2L^{2}$$

$$Z(t) = \frac{1}{\beta} \left[ Z(t-1) - g(t)h'(t)^H + \epsilon(t)g(t)^H \right]$$
(40) 
$$e'(t) = \frac{1}{\|v(t)\|} x(t) - W(t-1)y'(t)$$
(46) 
$$NL + N$$

$$\mathbf{W}(t) = \mathbf{W}(t-1) + \mathbf{e}'(t)\mathbf{g}(t)^{H}$$

$$(44) \quad NL$$

## IV. PSEUDO-CODE AND PERFORMANCE ANALYSIS

The computational cost for each of the detailed pseudo-code of the MFAPI is presented in Table 1. As depicted in Table 1, the computational complexity of MFAPI is approximately (3NL + 2N). Hence, it is in the low-complexity class of algorithms that requires O(NL) operations. Compared to the computational costs of the FAPI in [10, Table 3], the MFAPI algorithm has the similar computation complexity to the FAPI.

<sup>\*</sup> The values of  $\boldsymbol{W}(0)$  and  $\boldsymbol{Z}(0)$  should be assigned properly in order to have a fast convergence rate. In general, W(0) should be orthonormal and  $\mathbf{Z}(0)$  be positive definite.

The step without equation number can be found in the paper.



In summary, we emphasize that, from the implementation point of view, the MFAPI is simpler and more stable than FAPI. This is mainly because that the two algorithms realize orthonormalization of  $\boldsymbol{W}(t)$  in different ways.

## A. CONVERGENCE RATE ANALYSIS

As mentioned in the introduction section, almost all subspace tracking algorithms can be categorized into two types based on the way of their subspace tracking techniques are implemented. In this section, a comparison on the convergence rate of the both types is presented. The first kind of subspace tracking algorithms, that can only track single subspace basis, have similar form of the power iteration algorithm as mentioned in Sect. III-B. The second kind of methods, which can simultaneously track dual subspace basis matrix with a simple sign change, can be expressed as:

$$\mathbf{y}(t) = \mathbf{W}(t-1)^H \mathbf{x}(t), \tag{48}$$

$$\mathbf{M}(t) = \mathbf{W}(t-1) \pm \mu \mathbf{x}(t) \mathbf{y}(t)^{H}, \tag{49}$$

$$W(t) = \operatorname{orthonorm}\{M(t)\},$$
 (50)

where "+" represents that the algorithm can track signal subspace while "-" represents tracking noise subspace.  $\mu$  is the step size with small value for numerical stability and orthonorm{·} denotes the orthonormalization of M(t).

Each kind of subspace tracking algorithms has a different exponential convergence rate at the transient-state step. Here we do not analyze the performance of algorithms at the steady-state step and recommend the interested readers to read [23] for more detail information. As given in [7], the exponential convergence rates of the first and second types of subspace tracking methods can be expressed as  $(\lambda_{L+1}/\lambda_L)^n$  and  $((1+\mu\lambda_{L+1})/(1+\mu\lambda_L))^n$ , respectively, where  $\lambda_L > \lambda_{L+1}$ . When the algorithm converges to a steady-state point, we have  $\lim_{n\to\infty}(((1+\mu\lambda_{L+1})/(1+\mu\lambda_L))^n=0$  or  $\lim_{n\to\infty}(((1+\mu\lambda_{L+1})/(1+\mu\lambda_L))^n=0$ . Since  $\mu>0$  is very small and  $0\leq \lambda_{L+1}<\lambda_L$ , we conclude easily that the first kind has a faster convergence rate than the second kind. The results presented in the simulation section also agree with this analysis.

#### B. NUMERICAL STABILITY ANALYSIS

In this section, we analyze the numerical stability of MFAPI. We particularly focus on the deviations between the product  $W(t)^H W(t)$  and the identity matrix  $I_L$ . Similar analysis is also given in [17]. Now, assume that  $W(t)^H W(t) = I_L + \xi(t)$ , where  $\xi(t)$  is difference from  $I_L$ . Then, by means of  $W(t)^H e(t) = 0$  and  $W(t-1)^H W(t-1) = I_L + \xi(t-1)$ , it follows that (51), as shown at the bottom of this page. Detail derivations can be found in Appendix A.

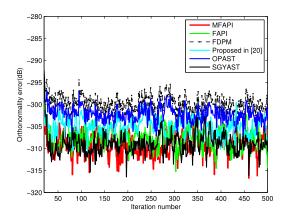


FIGURE 1. Orthonormality error analysis of MFAPI with other algorithms.

Assuming  $\boldsymbol{\xi}(t)$  is small at steady-state step (as seen in Fig. 1, the value of  $\|\boldsymbol{\xi}(t)\|_F^2$  is approximately  $10^{-30}$ ), it follows that  $\|\boldsymbol{v}(t)\|^2 + \frac{\boldsymbol{g}(t)^H}{\|\boldsymbol{g}(t)\|}\boldsymbol{\xi}(t-1)\frac{\boldsymbol{g}(t)}{\|\boldsymbol{g}(t)\|} \approx \|\boldsymbol{v}(t)\|^2$ . Consequently,

$$Y(t)^{H} T(t)^{H} T(t) Y(t) = \begin{bmatrix} \| \boldsymbol{v}(t) \|^{2} & \frac{\boldsymbol{g}(t)^{H}}{\| \boldsymbol{g}(t) \|} \boldsymbol{\xi}(t-1) \boldsymbol{G}(t) \\ \boldsymbol{G}(t)^{H} \boldsymbol{\xi}(t-1) \frac{\boldsymbol{g}(t)}{\| \boldsymbol{g}(t) \|} & \boldsymbol{I}_{L-1} + \boldsymbol{G}(t)^{H} \boldsymbol{\xi}(t-1) \boldsymbol{G}(t) \end{bmatrix}.$$
(52)

Equations (36), (47) and (52) yield

$$W(t)^{H}W(t) = \frac{g(t)g(t)^{H}}{\|g(t)\|^{2}} + \frac{1}{\|v(t)\|}G(t)G(t)^{H}\xi(t-1)\frac{g(t)g(t)^{H}}{\|g(t)\|^{2}} + \frac{g(t)g(t)^{H}}{\|v(t)\|\|g(t)\|^{2}}\xi(t-1)G(t)G(t)^{H} + G(t)G(t)^{H} + G(t)G(t)^{H}\xi(t-1)G(t)G(t)^{H}.$$
 (53)

Replacing  $G(t)G(t)^H = I_L - \frac{g(t)g(t)^H}{\|g(t)\|^2}$ , (53) can be rewritten as

$$W(t)^{H}W(t) = I_{L} + \xi(t-1) + (1 - \frac{2}{\|v(t)\|}) \frac{g(t)g(t)^{H}}{\|g(t)\|^{2}} \xi(t-1) \frac{g(t)g(t)^{H}}{\|g(t)\|^{2}} + (\frac{1}{\|v(t)\|} - 1) \frac{g(t)g(t)^{H}}{\|g(t)\|^{2}} \xi(t-1) + (\frac{1}{\|v(t)\|} - 1) \xi(t-1) \frac{g(t)g(t)^{H}}{\|g(t)\|^{2}}.$$
 (54)

$$Y(t)^{H}T(t)^{H}T(t)Y(t) = \begin{bmatrix} \|v(t)\|^{2} + \frac{g(t)^{H}}{\|g(t)\|}\xi(t-1)\frac{g(t)}{\|g(t)\|} & \frac{g(t)^{H}}{\|g(t)\|}\xi(t-1)G(t) \\ G(t)^{H}\xi(t-1)\frac{g(t)}{\|g(t)\|} & I_{L-1} + G(t)^{H}\xi(t-1)G(t) \end{bmatrix}.$$
(51)



Substituting  $\mathbf{W}(t)^H \mathbf{W}(t) = \mathbf{I}_L + \boldsymbol{\xi}(t)$  into (54) yields

$$\xi(t) = \xi(t-1) + (1 - \frac{2}{\|\boldsymbol{v}(t)\|}) \frac{\boldsymbol{g}(t)\boldsymbol{g}(t)^{H}}{\|\boldsymbol{g}(t)\|^{2}} \xi(t-1) \frac{\boldsymbol{g}(t)\boldsymbol{g}(t)^{H}}{\|\boldsymbol{g}(t)\|^{2}} + (\frac{1}{\|\boldsymbol{v}(t)\|} - 1) \frac{\boldsymbol{g}(t)\boldsymbol{g}(t)^{H}}{\|\boldsymbol{g}(t)\|^{2}} \xi(t-1) + (\frac{1}{\|\boldsymbol{v}(t)\|} - 1) \xi(t-1) \frac{\boldsymbol{g}(t)\boldsymbol{g}(t)^{H}}{\|\boldsymbol{g}(t)\|^{2}}.$$
 (55)

With reference to the similar derivation in [17], we discuss the numerical stability in two cases in the presence of different SNR values:

Case 1: In high SNR region, since all the noise eigenvalues of sample covariance matrix are sufficiently small,  $\|e(t)\|^2$  becomes sufficiently small (according to Karhunen-Loève transform (KLT), the error of projection  $\|e(t)\|^2 = \sum_{k=L+1}^{N} \lambda_k$  is used). From Equation (35) then it follows that  $\|v(t)\| \approx 1$  and hence (55) can be rewritten as

$$\xi(t) = \xi(t-1) - \frac{g(t)g(t)^{H}}{\|g(t)\|^{2}} \xi(t-1) \frac{g(t)g(t)^{H}}{\|g(t)\|^{2}} + \Gamma(t),$$
(56)

where  $\Gamma(t)$  is the numerical precision error and here we assume  $\Gamma(t) = 0$ . Applying the column-wise version  $\text{vec}\{\xi(t)\}$  of  $\xi(t)$ , (56) takes the form

$$\operatorname{vec}\{\boldsymbol{\xi}(t)\} = \mathbf{\Omega}\operatorname{vec}\{\boldsymbol{\xi}(t-1)\},\tag{57}$$

where  $\Omega = I_{L^2} - \frac{g(t)g(t)^H}{\|g(t)\|^2} \otimes \frac{g(t)g(t)^H}{\|g(t)\|^2}$  and  $\otimes$  denotes the Kronecker product. In order to achieve the numerical stability, we need  $\lim_{t\to\infty} \mathrm{vec}\{\xi(t)\} = 0$  in (57). In this case all eigenvalues of matrix  $\Omega$  should be in the unit circle. Since  $\frac{g(t)g(t)^H}{\|g(t)\|^2}$  is positive definite and  $\mathrm{tr}\{\frac{g(t)g(t)^H}{\|g(t)\|^2}\} = 1$ , the eigenvalues,  $\gamma_i(i=1,2,...,L)$ , of  $\frac{g(t)g(t)^H}{\|g(t)\|^2}$  satisfy  $0<\gamma_i<1$  (i=1,2,...,L). Meanwhile, by using properties of the Kronecker product eigenvalues, we can show the eigenvalues of  $\Omega$  satisfying

$$0 < (1 - \gamma_i \gamma_i) < 1, \quad i, j = 1, 2, ..., L.$$
 (58)

Case 2: In low SNR region,  $\|e(t)\|^2$  is not sufficiently small. However, according to Fig. 1,  $\|\xi(t)\|_F^2$  is small enough to be taken approximately equal to zero. Hence, the resulting derivation becomes very simple by directly taking  $\xi(t-1)=0$ . In this case, (55) can be written as  $\xi(t)=\xi(t-1)\approx 0$ .

Consequently, from the above discussions, we conclude that the MFAPI is numerically stable.

#### V. SIMULATION RESULTS

In this section, computer simulation results are presented to test the performance of the proposed MFAPI algorithm. We note that the subspace tracking algorithms are extremely rich in the literature and hence it is difficult to assess and compare the performances of all the algorithms. In order to have a relatively complete comparison, five well-known

algorithms (FAPI, OPAST, FDPM, SGYAST, and algorithm proposed in [20]) are used for comparison. Among those algorithms, FAPI and OPAST belong to the first type while FDPM and the algorithm in [20] belong to the second type (see the introduction section). As a stable version of GYAST, SGYAST, is included since this method is newer in the literature despite its higher computational complexity O(6NL) than that of other algorithms O(3NL) (here we do not consider GYAST as a test bench due to its numerical instability, as discussed in [19]). We compare the performance of these algorithms from two aspects: orthonormality error and convergence rate.

### A. ORTHONORMALITY ANALYSIS

Performance of subspace tracking methods can be measured in part by orthonormality error of W(t) and thus the following criteria is commonly used to estimate this error.

$$10\log_{10} \|\boldsymbol{W}(t)^{H} \boldsymbol{W}(t) - \boldsymbol{I}_{L}\|_{F}^{2}. \tag{59}$$

Fig. 1 shows the orthonormality errors of OPAST, FAPI, MFAPI, SGYAST, FDPM, and algorithm proposed in [20]. We observe that these algorithms all have superior performance for orthonormality and the value reduces to about -300 dB.

TABLE 2. Parameters used in subspace tracking methods.

Parameters	Value	Algorithms
Frequencies*	0.1Hz,0.2Hz,0.3Hz	all
Sample Number	80	all
SNR	15dB	all
$\beta$	0.87	FAPI, MFAPI, OPAST,SGYAST
$\mu$	0.009	FDPM, Proposed in [20]

<sup>\*</sup>The abrupt frequency change of 0.3Hz is at iteration point 100 and 0.2Hz at iteration point 400, as shown in Fig.2.

## **B. CONVERGENCE RATE ANALYSIS**

In this section, to test the convergence rates of the MFAPI and FAPI, we consider estimation of the frequencies of signals embedded in additive noise. We assume that frequencies of three sinusoidal signals are estimated with the rank of signal subspace basis matrix to be L = 3. Additionally, in order to improve the performance of the algorithm, Root-MUSIC method is also applied to the estimated subspace basis matrix [24]. The SNR is chosen approximately 15dB as in [17]. More detail configurations are also shown in Table 2. In order to observe the convergence ability during sudden steep rises or drops in frequencies, we set them change abruptly at iteration points 100 and 400. Since MFAPI is a modified version of FAPI, we compare the behavior of these two algorithms first. As seen from the signal tracking performance of MFAPI and FAPI in Fig. 2, both of these two algorithms having capability to track the frequencies of signals, even in the presence of abrupt situations. As Fig. 2 provides an overall description, they have similar performances in the



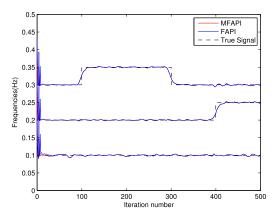


FIGURE 2. Tracking three signals with MFAPI and FAPI algorithm.

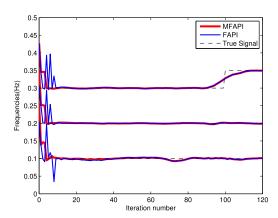


FIGURE 3. Tracking three signals with MFAPI and FAPI for local enlargement.

context of signal detection and tracking. However, a more clear differences between these algorithms can be observed by a local enlargement, as seen in Fig. 3, where the MFAPI has a faster convergence rate than FAPI at the transient-state stage.

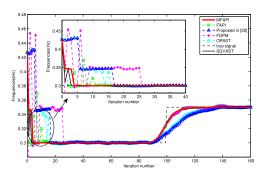


FIGURE 4. Tracking one signal with different algorithms.

Unlike Fig. 3, plotting the tracking curves of all the three frequencies, Fig. 4 provides only a signal with frequency 0.3 Hz for explicit comparison. The MFAPI and other five algorithms (FAPI, OPAST, FDPM, SGYAST, and algorithm proposed in [20]) are chosen as test benches and it is evident that the proposed MFAPI method has a faster convergence rate than all other algorithms at transient-state steps except SGYAST, as depicted in Fig. 4 (Please note although SGYAST has slightly better performance than MFAPI, SGYAST has higher computation complexity). The results clearly verify the conclusion of the analysis, presented in Section IV.

## VI. CONCLUSION

In this paper, a fast and stable subspace tracking algorithm with low complexity is provided. The proposed algorithm has the similar computational complexity as FAPI. Instead of solving the inverse square root of matrix as employed in FAPI, the proposed algorithm applies direct matrix product to ensure the orthonormality of subspace basis matrix at each recursion, that has more simpler derivation. Additionally, the new algorithm is proven to be numerically stable as shown by the detail mathematical inference. Finally, regarding the convergence rate at transient-state step, our improved

$$Y(t)^{H}T(t)^{H}T(t)Y(t) = \begin{bmatrix} \frac{g(t)^{H}}{\|g(t)\|} W(t-1)^{H}W(t-1) + \|g(t)\|e^{H}(t)e(t)g(t)^{H} \\ G(t)^{H}W(t-1)^{H}W(t-1) \end{bmatrix} \begin{bmatrix} \frac{g(t)}{\|g(t)\|} \\ \frac{g(t)}{\|g(t)\|} \end{bmatrix} G(t)$$

$$= \begin{bmatrix} \frac{g(t)^{H}}{\|g(t)\|} W(t-1)^{H}W(t-1) \frac{g(t)}{\|g(t)\|} + \|g(t)\|^{2} \|e(t)\|^{2} \frac{g(t)^{H}}{\|g(t)\|} W(t-1)^{H}W(t-1)G(t) \\ G(t)^{H}W(t-1)^{H}W(t-1) \frac{g(t)^{H}}{\|g(t)\|} & G(t)^{H}W(t-1)^{H}W(t-1)G(t) \end{bmatrix}$$

$$Y(t)^{H}T(t)^{H}T(t)Y(t) = \begin{bmatrix} \frac{g(t)^{H}}{\|g(t)\|} \xi(t-1) \frac{g(t)}{\|g(t)\|} + 1 + \|g(t)\|^{2} \|e(t)\|^{2} & \frac{g(t)^{H}}{\|g(t)\|} \xi(t-1)G(t) \\ G(t)^{H}\xi(t-1) \frac{g(t)}{\|g(t)\|} & I_{L-1} + G(t)^{H}\xi(t-1)G(t) \end{bmatrix}$$

$$= \begin{bmatrix} \|v(t)\|^{2} + \frac{g(t)^{H}}{\|g(t)\|} \xi(t-1) \frac{g(t)}{\|g(t)\|} & \frac{g(t)^{H}}{\|g(t)\|} \xi(t-1)G(t) \\ G(t)^{H}\xi(t-1) \frac{g(t)}{\|g(t)\|} & I_{L-1} + G(t)^{H}\xi(t-1)G(t) \end{bmatrix}$$

$$(63)$$



algorithm yields the fastest convergence rate among existing subspace tracking methods with computation complexity O(3NL) as verified by numerical simulations.

## **APPENDIX I. DERIVATION OF EQUATION (51)**

Equations (31), (33) yield

$$Y(t)^{H}T(t)^{H}$$

$$=\begin{bmatrix} \frac{g(t)^{H}}{\|g(t)\|} \\ \dots \\ G(t)^{H} \end{bmatrix} [W(t-1)^{H} + g(t)e(t)^{H}]$$

$$=\begin{bmatrix} \frac{g(t)^{H}}{\|g(t)\|}W(t-1)^{H} + \|g(t)\|e(t)^{H} \\ G(t)^{H}W(t-1)^{H} \end{bmatrix}$$
(60)

Equation (60) post-multiplied by T(t) yields

$$Y(t)^{H}T(t)^{H}T(t)$$

$$= \begin{bmatrix} \frac{g(t)^{H}}{\|g(t)\|} W(t-1)^{H} + \|g(t)\|e(t)^{H} \\ G(t)^{H}W(t-1)^{H} \end{bmatrix} [W(t-1) + e(t)g(t)^{H}]$$

$$= \begin{bmatrix} \frac{g(t)^{H}}{\|g(t)\|} W(t-1)^{H}W(t-1) + \|g(t)\|e(t)^{H}e(t)g(t)^{H} \\ G(t)^{H}W(t-1)^{H}W(t-1) \end{bmatrix}$$
(61)

The next step derivation, Equation (62), can be found at the bottom of the previous page.

By using  $W(t-1)^H W(t-1) = I_L + \xi(t-1)$  we obtain (63), as shown at the bottom of the previous page.

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