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MXPFIT: A library for finding optimal multi-exponential approximations*



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ABSTRACT

mxpfit is a library implemented in C++ to find optimal approximations of functions by multi-exponential sums with complex-valued parameters. The library provides an interface for evaluating exponents and coefficients from sampling data on a uniform grid using the fast Estimation of Signal Parameters via Rotational Invariance Techniques (ESPRIT) algorithm originally proposed by Potts and Tasche (2015). The parameters can be estimated efficiently from a sampling data even including noise. A modified balanced truncation algorithm to find the multi-exponential sum with a smaller order is also provided. These features are useful for finding optimal exponential sum approximations of analytic functions or large-scale numerically sampled data set.

Program summary

Program Title: mxpfit

Program Files doi: http://dx.doi.org/10.17632/vjygkwpss4.1

Licensing provisions: MIT Programming language: C++

Nature of problem: Nonlinear curve fitting by multi-exponential function, frequency estimation of signals, and reduction of multi-exponential sum

Solution method: The modified fast Estimation of Signal Parameters via Rotational Invariance Techniques (ESPRIT) algorithm originally proposed in [1] is utilized for the parameter estimation of multi-exponential sums from sampled data on a uniform grid. In this method, partial Lanczos bidiagonalization is used to find a low-rank approximation of a Hankel matrix constructed from sampled data. An overdetermined Vandermonde system is solved by the conjugate gradient least square (CGLS) method with preconditioner to obtain the coefficients of the multi-exponential sum. The truncation of the multi-exponential sum is performed by a modified balanced truncation method in [2]. The projection matrix to obtain the truncated system is determined by computing con-eigenvalues of a Gramian matrix, which can be computed efficiently with highly relative accuracy using the algorithm in [3].

Additional comments: A C++ compiler supporting the C++11 standard is required to compile programs using the mxpfit library. mxpfit depends on the Eigen (http://eigen.tuxfamily.org/) and FFTW3 (http://www.fftw.org/) libraries. CMake (https://cmake.org/) is used for building example programs. References:

- [1] D. Potts and M. Tasche, Appl. Numer. Math. 88 (2015) 31.
- [2] K. Xu and S. Jiang, J. Sci. Comput. 55 (2013) 16.
- [3] T. S. Haut and G. Beylkin, SIAM J. Matrix Anal. Appl. 33 (2013) 1101.

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1. Introduction

We consider functions expressed as a sum of exponential functions given as

$$f(t) = \sum_{i=1}^{M} c_i e^{-a_i t} \quad (t \ge 0)$$
 (1)

[↑] This paper and its associated computer program are available via the Computer Physics Communication homepage on ScienceDirect (http://www.sciencedirect.com/science/journal/00104655).

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where $a_j \in \mathbb{C}$, $\operatorname{Re}(a_j) > 0$ are distinct complex numbers, and $c_j \in \mathbb{C} \setminus \{0\}$. The function f(t) is a smooth, exponentially decaying function, and it oscillates if $\operatorname{Im}(a_j) \neq 0$. Recovering the parameters in Eq. (1) from noisy sampled data is a common problem in the field of signal processing. Exponential data fitting was already applied in electrical engineering and in many mathematical physics applications, e.g., fitting data in magnetic resonance imaging (MRI) and modeling spectroscopic data [1].

Prony's method is a well-known algorithm for finding the best M-term approximation of f(t) from uniformly sampled signals. However, this method may be unstable, especially when the case signal is embedded in noise. Various modifications of Prony's method to improve its numerical stability have been proposed [2,3]. Among them, the Estimation of Signal Parameters via Rotational Invariance Techniques (ESPRIT) method and its variants are commonly used for their robustness against noise [4,5]. In the original ESPRIT method, we need a singular value decomposition (SVD) of the Hankel matrix constructed from sampled data. Fast decomposition algorithms relying on the special structure of the Hankel matrix are available, and were used in the fast ESPRIT method [6].

Approximating analytic functions by exponential sums has also been studied in applied mathematics. Beylkin and Monzón proposed a Prony-like method for finding the best approximation of analytic functions on finite intervals from function values on equally spaced mesh points [7]. This method is based on the theory of Adamjan, Arov, and Krein (AAK theory). Andersson et al. made further mathematical considerations for this method [8]. They also proposed a similar algorithm (the so-called fast alternating projection method) for parameter estimation [9].

Bevlkin and Monzón introduced another approach to obtain exponential sum approximations of analytic functions by discretizing the integral representations using a numerical quadrature rule [10]. The order of the exponential sum obtained by this procedure is usually suboptimal. The modified Prony's method for obtaining the optimal number of terms was also discussed in this literature. Alternatively, balanced truncation, which is an algorithm used for model order reduction (MOR), can also be used to find optimal approximations by exponential sums [11,12]. These techniques are suitable for approximating functions with singularities, such as power functions $r^{-\beta}$ ($\beta>0$). This approximation was applied to construct Green's functions of Laplace equations appearing in quantum chemistry and fluid dynamics [13–15]. For example, Genovese et al. developed a fast Laplace equation solver that was used to evaluate integrals for electron-electron interaction energies [16]. The exponential sum approximation is also useful for efficient evaluation of integral transformation with an oscillatory kernel [17].

The mxpfit library was developed for numerically computing the exponential sum approximation of functions efficiently and with controllable accuracy. This library provides application programming interfaces (APIs) for (1) recovering parameters in Eq. (1) from sampled data on a uniform grid, and (2) reducing the order (number of terms) M of a given exponential sum. The modified fast ESPRIT algorithm was adopted for (1). A fast least squares solver for a rectangular Vandermonde matrix, which is required for computing coefficients $\{c_i\}$, was also developed. The balanced truncation method specialized for an exponential sum was designed and implemented for (2). The con-eigensolver of a quasi-Cauchy matrix developed by Haut and Beylkin was adopted to execute the truncation efficiently and with high relative accuracy [18].

Recently, the present author has applied this library to evaluate the spherical Bessel transform (SBT), which is defied as,

$$F(k) = \int_0^\infty r^2 f(r) j_l(kr) dr, \tag{2}$$

where the $j_l(r)$ is the spherical Bessel function. This can be evaluated efficiently and accurately if exponential sum approximation of residual function $r^2f(r)$ and oscillating kernel $j_l(r)$ are obtained in high-accuracy. The details about the procedure can be found elsewhere [19].

2. Fast ESPRIT method

In this section, a fast ESPRIT algorithm to estimate the parameters a_i , c_i and number of terms M of exponential sums in Eq. (1) is briefly introduced. A detailed description of the fast ESPRIT algorithm can be found in the literature [6]. For the mathematical background about the ESPRIT and other Prony-like methods, we refer to [2,3,7].

Let $\mathbf{f} = (f_0, f_1, \dots, f_{N-1})^T$ be the values of f(t) sampled on a uniform grid $t_k = t_0 + hk$, $(h > 0, k = 0, 1, \dots, N-1)$ with a sufficiently large number of $N \ge 2M$. Each element f_k can be written as

$$f_k = \sum_{i=1}^{M} c_j e^{-a_j(t_0 + hk)} = \sum_{i=1}^{M} w_i z_j^k,$$
(3)

where $z_j = e^{-a_j h}$ and $w_j = c_j e^{-a_j t_0}$. The problem of obtaining the best exponential sum approximation of the form (1) becomes the goal of finding complex weights $\mathbf{w} = (w_0, \dots, w_M)^T$ and complex nodes $\mathbf{z} = (z_1, \dots, z_M)^T$ with a minimal number of terms M. Then, we introduce a rectangular Hankel matrix

$$H = \begin{pmatrix} f_0 & f_1 & \cdots & f_{K-1} \\ f_1 & f_2 & \cdots & f_K \\ \vdots & \vdots & \ddots & \vdots \\ f_{L-1} & f_L & \cdots & f_{L+K-2} \end{pmatrix} \in \mathbb{C}^{L \times K}, \tag{4}$$

where $M \le L \le N-M+1$ and K=N-L+1. Note that $L \ge K$. In the data processing application, the sequence \boldsymbol{f} is called a time series of length N, the matrix H is called a L-trajectory matrix with a window length of L. The Hankel matrix H is usually singular when N is sufficiently large. Prony-like methods use the fact that the rank of H coincides with the number of terms M in Eq. (3). The ESPRIT method uses the singular value decomposition (SVD) of the Hankel matrix

$$H = U\Sigma W^*, \tag{5}$$

where $U \in \mathbb{C}^{L \times L}$ and $W \in \mathbb{C}^{K \times K}$ are unitary matrices, and Σ is the diagonal matrix holding the singular values as diagonal elements in ascending order $\sigma_1 \geq \sigma_2 \geq \cdots \geq \sigma_M > \sigma_{M+1} = \cdots = \sigma_K = 0$. In practice, the rank M is determined so that $\sigma_{M+1} < \epsilon$, where $\epsilon > 0$ is the required accuracy for the approximation. The nodes z_j are computed as the eigenvalues of the matrix

$$F_{\text{SVD}} = W_M^{\dagger}(0)W_M(1) \in \mathbb{C}^{M \times M}, \tag{6}$$

where the matrices $W_M(0)$ and $W_M(1)$ are defined as $W_M(0) = W[1:K-1,1:M]$ and $W_M(1) = W[2:K,1:M]$, respectively, and the dagger (\dagger) indicates the Moore–Penrose pseudoinverse of the matrix.

One of the drawbacks in the original ESPRIT method is the high computational cost of order $O(N^3)$ for computing the full SVD of the Hankel matrix. This makes it difficult to apply the method to a large number of sample data. In the fast ESPRIT algorithm, a partial Lanczos bidiagonalization is utilized to obtain a low-rank approximation of the Hankel matrix (4).

The Golub–Kahan–Lanczos bidiagonalization decomposes the general matrix $A \in \mathbb{C}^{L \times K}$ in the form

$$A = P_K B_K Q_K^*, \tag{7}$$

where $P_n = [\boldsymbol{p}_1, \dots, \boldsymbol{p}_n] \in \mathbb{C}^{L \times n}$ and $Q_n = [\boldsymbol{q}_1, \dots, \boldsymbol{q}_n] \in \mathbb{C}^{K \times n}$ are matrices with orthonormal columns, and B_n is the bidiagonal real matrix

$$B_{n} = \begin{pmatrix} \alpha_{1} & \beta_{1} & & & & \\ & \alpha_{2} & \beta_{2} & & & \\ & & \ddots & \ddots & & \\ & & & \alpha_{K-1} & \beta_{K-1} \\ & & & & \alpha_{K} \end{pmatrix} \in \mathbb{R}^{K \times K}.$$
 (8)

The column vectors \mathbf{p}_j and \mathbf{q}_j are called the left and right Lanczos vectors, respectively. The bidiagonal elements α_i , β_i and the Lanczos vectors are obtained using the recurrence relation

$$\alpha_1 \mathbf{p}_1 = A \mathbf{q}_1,$$

$$\alpha_{j+1} \mathbf{p}_{j+1} = A \mathbf{q}_{j+1} - \beta_j \mathbf{p}_j,$$

$$\beta_i \mathbf{q}_{j+1} = A^* \mathbf{p}_i - \alpha_i \mathbf{q}_i,$$
(9)

starting from an arbitrary selected unit vector \mathbf{q}_1 . When computing the recurrence (9) in finite precision arithmetic, the orthogonality between the Lanczos vectors might be lost. In order to recover the orthogonality, several reorthogonalization schemes have been proposed. See Ref. [20] for more details about the Lanczos method and the reorthogonalization scheme. In the present implementation of mxpfit, the full-reorthogonalization scheme was adopted following [6] for numerical robustness.

Let us consider the matrix $\tilde{A}_M = P_M B_M Q_M^*$, where P_M , Q_M and P_M are obtained by applying the bidiagonalization step P_M times ($P_M \leq K$). As discussed in [6], P_M and P_M contain good approximations of the singular vectors related to the first P_M dominant singular values. Hence, P_M could be regarded as a low-rank approximation of matrix P_M . The approximation error of P_M with respect to P_M is defined in the Frobenius norm as $P_M = ||P_M||_F$. The approximation error can be determined recursively as

$$\Delta_{M+1}^2 = \Delta_M^2 - \alpha_{M+1}^2 - \beta_M^2, \ (2 \le M \le K - 1)$$

$$\Delta_1^2 = ||A||_F - \alpha_1^2, \tag{10}$$

and $\Delta_K = 0$. In addition,

$$\|\tilde{A}_{M}\|_{F}^{2} = \|B_{M}\|_{F}^{2} = \alpha_{1}^{2} + \sum_{i=1}^{M-1} (\alpha_{j+1}^{2} + \beta_{j}^{2})$$
(11)

converges monotonically to $\|A\|_F^2$. Thus, we can utilize the magnitude of $\alpha_{M+1}^2 + \beta_M^2$ as an estimator of the approximation error. In practice, we monitor the value in each bidiagonalization step and stop the recurrence if the condition $\alpha_{M+1}^2 + \beta_M^2 < \epsilon^2 \|\tilde{A}_M\|_F^2$ is met.

The low-rank approximation of the Hankel matrix $H \in \mathbb{C}^{L \times K}$ is computed using the partial Lanczos bidiagonalization described above. In this process, two Hankel matrix-vector multiplication operations are required in each bidiagonalization step. The operation can be performed efficiently using a fast Fourier transform (FFT) [6,21]. The algorithm can be written as follows:

Algorithm 1 (General Fast Hankel Matrix–Vector Product). Input: $L, K \in \mathbb{N}$ and a sequence $\boldsymbol{f} = (f_0, \dots, f_{L+K-1})^T$ to form a $L \times K$ Hankel matrix $H, P \in \mathbb{N}$ the length of the FFT with $P \geq L + K - 1$, and a vector \boldsymbol{x} of length K (\boldsymbol{x}' of length L). Output: $\boldsymbol{y} := H\boldsymbol{x}$ of length L ($\boldsymbol{y}' := H^*\boldsymbol{x}'$ of length K).

1. Form the vector $\tilde{\mathbf{f}}$ of length P

$$\tilde{\boldsymbol{f}} = (f_{K-1}, f_K, \dots, f_{L+K-2}, \underbrace{0, \dots, 0}_{P-L-K+1}, f_0, f_1, \dots, f_{K-2}).$$

2. Form the vector $\tilde{\mathbf{x}}$ of length P

$$\tilde{\mathbf{x}} = (x_{K-1}, x_{K-2}, \dots, x_0, \underbrace{0, \dots, 0}_{P-K}) \qquad \text{for multiplying } H$$

$$\tilde{\mathbf{x}} = (\underbrace{0, \dots, 0}_{P-L}, x_{L-1}^{\prime*}, x_{L-2}^{\prime*}, \dots, x_0^{\prime*}) \qquad \text{for multiplying } H^*$$

3. Compute the vector $\tilde{\mathbf{v}}$ of length P as

$$\tilde{\mathbf{v}} := \text{IFFT}(\text{FFT}(\tilde{\mathbf{f}}) * \text{FFT}(\tilde{\mathbf{x}}))$$

where '*' denotes an element-wise multiplication of two vectors.

4. Form the vector $\mathbf{y}(\mathbf{y}')$ as

$$\mathbf{y} = \tilde{\mathbf{y}}[0:L-1]$$
 for multiplying H
 $\mathbf{y}' = \tilde{\mathbf{y}}[P-K:P-1]$ for multiplying H^* .

Here, the FFT and IFFT in Algorithm 1 denote a one-dimensional FFT and inverse FFT, respectively. In the present case, steps 2–4 are repeated 2M times during the M bidiagonalization step with f fixed. Therefore, once the vector \tilde{f} in step 1 of Algorithm 1 is constructed, FFT(\tilde{f}) is computed and stored in a temporary vector so that we can reuse it in step 3. The computational cost of the Hankel matrix–vector operation is $O(P \log P)$ per single product, while that for the general matrix–vector product is O(LK). Note that the optimal FFT size is in the range $L + K - 1 \le P < 2(L + K - 1)$. The computational complexity for the M-step Lanczos bidiagonalization for the Hankel matrix becomes $O(MP \log P)$, which is much smaller than that of the full SVD.

Once a low-rank approximation of the Hankel matrix $H = P_M B_M Q_M^*$ is obtained, the nodes z_j are computed as the eigenvalues of the matrix

$$F_{\text{LBD}} = Q_{M}^{\dagger}(0)Q_{M}(1) \in \mathbb{C}^{M \times M}, \tag{12}$$

where $Q_M(0) = Q_M[1:K-1,1:M]$ and $Q_M(1) = Q_M[2:K,1:M]$, respectively. The computation of the Moore-Penrose pseudoinverse Q_M^{\dagger} is expensive because it requires computing the SVD of Q_M . In the original paper of the fast ESPRIT method, this step is bypassed by solving a generalized eigenvalue problem yielding the same eigenvalues. Badeau et al., used a different method of computing F_{LBD} without explicitly computing the pseudoinverse as

$$F_{\text{LBD}} = \Psi + \frac{1}{1 - \|\mathbf{v}\|_{2}^{2}} \mathbf{v} (\Psi^{*} \mathbf{v})^{*}$$
(13)

with $\Psi = Q_M^*(0)Q_M(1)$ and $v = Q_M[K, 1:M]$ [22]. This method relies on the orthonormality between column vectors \mathbf{q}_j of matrix Q_M . The mxpfit library adopted the latter approach.

After obtaining nodes z_j , weights w_j in (3) are computed as the least squares solution of the linear system:

$$V(z)w = f, (14)$$

where V(z) is a rectangular Vandermonde matrix defined by nodes z as

$$V(\mathbf{z}) = \begin{pmatrix} 1 & 1 & \cdots & 1 \\ z_1 & z_2 & \cdots & z_M \\ z_1^2 & z_2^2 & \cdots & z_M^2 \\ \vdots & \vdots & \ddots & \vdots \\ z_1^{N-1} & z_2^{N-1} & \cdots & z_M^{N-1} \end{pmatrix} \in \mathbb{C}^{N \times M}.$$
 (15)

There is little discussion in literature about the computational complexity for solving this overdetermined Vandermonde system (14) related to parameter estimation. The least squares problem for the complex general matrix can be solved using subroutine ZGELSD implemented in the linear algebra package (LAPACK). This

subroutine internally uses the SVD of the matrix. The computational cost for the SVD of $N \times M$ matrix V(z) is $O(N^2M + NM^2 + M^3)$. Although the number of terms M is usually small, the computational time for solving (14) becomes significant if the number of sample data N is large.

In the present work, the conjugate gradient least square (CGLS) method was adopted to solve the overdetermined Vandermonde system (14). The convergence of the CGLS method can be accelerated if an appropriate preconditioner, which approximately solves a linear system

$$V^*(z)V(z)x = V^*(z)b, \tag{16}$$

is provided. In the present work, we designed a preconditioner using the fast decomposition algorithm for a Vandermonde matrix. In practice, the Cholesky decomposition

$$V^*(\mathbf{z})V(\mathbf{z}) = X\Sigma^2 X^* \tag{17}$$

is computed, where $X \in \mathbb{C}^{M \times M}$ is a unit lower triangular matrix, and $\Sigma \in \mathbb{R}^{M \times M}$ is a diagonal matrix with nonnegative entries. The preconditioning linear system can be solved as

$$\mathbf{x} = X^{-*} \Sigma^{-2} X^{-1} (V^*(\mathbf{z}) \mathbf{b}), \tag{18}$$

with $O(M^2)$ computational cost. The computational complexity of the Cholesky decomposition (17) is only $7M^2/2 + O(n)$, using the algorithm proposed by Demeure (Algorithm C in Ref. [23]). The least squares solution of Eq. (14) can usually be obtained with a small number of iterations in CGLS (fewer than 10 in most cases). Each CG step requires two matrix–vector products, which cost O(NM) for each product. Thus, the total complexity of the new Vandermonde least squares solver is $O(NM+M^2)$, which is significantly smaller than that for a conventional least squares solver using the SVD. Note that this CGLS algorithm might not converge if ϵ is too small. In such a case, the present code automatically falls back to the standard least square algorithm using the SVD to achieve the correct weights.

Once nodes z_j and w_j are computed, the parameters a_j and c_j for the exponential sum can be obtained as $a_j = -(\log z_j)/h$ and $c_j = w_j e^{-a_j t_0}$. Our modified fast ESPRIT algorithm is summarized in Algorithm 2.

Algorithm 2 (*Modified Fast ESPRIT Method*). Input: Sequence of function values f_k sampled on an equispaced grid $t_k = t_0 + kh$ for $k = 0, 1, \ldots, N-1$, the window length $L \le N$, M_b the upper bound of number of terms M, and prescribed accuracy $\epsilon > 0$. Output: parameters a_j with $\operatorname{Re}(a_j) > 0$ and c_j $(j = 1, \ldots, M)$ for the exponential $\operatorname{sum} f(t)$ such that $||f(t_k) - f_k|| < ||f||_2 \epsilon$

- 1. Compute M-step partial Lanczos bidiagonalization of the Hankel matrix H in (4) such that $\|H P_M B_M Q_M^*\| < \epsilon$. All Hankel matrix-vector products are computed using Algorithm 1.
- 2. Form matrix F_{LDB} using (13), and compute its eigenvalues z_j (j = 1, ..., M). Set $a_j = -(\log z_j)/h$ (j = 1, ..., M).
- 3. Compute w_j ($j=1,\ldots,M$) as a least squares solution of the overdetermined Vandermonde system (14) using a CGLS solver with a preconditioner (18) (automatically fall back to the least squares method with SVD if not converged). Set $c_j = w_j e^{-a_j t_0}$ ($j=1,\ldots,M$).

${\bf 3.} \ \ {\bf Balanced} \ truncation \ algorithm \ for \ exponential \ sum$

For a prescribed accuracy $\epsilon>0$, we consider the approximation of f(t) with another exponential sum with a smaller number of terms, such that

$$\hat{f}(t) = \sum_{j=1}^{M'} \hat{c}_j e^{-\hat{a}_j t}, \quad \|f(t) - \hat{f}(t)\| < \epsilon, \tag{19}$$

where $M' \leq M$. Let us consider the Laplace transform of f(t), which can be evaluated analytically as

$$F(s) = \mathcal{L}[f(t)] = \int_0^\infty e^{-st} f(t) dt = \sum_{i=1}^M \frac{c_i}{s + a_i}.$$
 (20)

Thus, the exponents a_i and coefficients c_i are the poles and residues of the Laplace transform F(s). The problem of finding an optimal exponential sum (19) becomes a problem of finding a shorter sum-of-poles approximation of (20), such that

$$\hat{F}(s) = \sum_{i=1}^{M'} \frac{\hat{c}_i}{s + \hat{a}_i}, \ \|F(s) - \hat{F}(s)\| < \epsilon$$
 (21)

for all s > 0.

Let $A = \text{diag}(a_1, a_2, \dots, a_M)$, $B = (\sqrt{c_1}, \sqrt{c_2}, \dots, \sqrt{c_M})^T$, and $C = B^T$. Then, f(t) and F(s) can be expressed as

$$f(t) = Ce^{-tA}B, (22)$$

and

$$F(s) = C(sI + A)^{-1}B, (23)$$

respectively. Using the terminology of *control theory*, F(s) is a transfer function of a linear time-invariant (LTI) system, which is described by a differential equation given as

$$\begin{aligned} \frac{d\mathbf{x}(t)}{dt} &= -A\mathbf{x}(t) + Bu(t), \ \mathbf{x}(0) = \mathbf{0} \\ y(t) &= C\mathbf{x}(t), \end{aligned} \tag{24}$$

where u(t) is the control (input), y(t) is the output, and $x(t) = (x_1(t), \ldots, x_M(t))$ are the so-called state variables. By applying the Laplace transform to Eq. (24), we obtain

$$sX(s) = -AX(s) + BU(s),$$

$$Y(s) = CX(s).$$
(25)

The transfer function F(s) defines the linear mapping between the Laplace transform of the input $U(s) = \mathcal{L}[u(t)]$ and that of the output $Y(s) = \mathcal{L}[y(t)]$ as Y(s) = F(s)U(s). Approximating such a dynamic system with a smaller number of state variables is a common problem of the MOR in control theory [24].

Among many techniques of MOR, balanced truncation is a common technique to approximate an LTI system using projection [25]. This method relies on the fact that applying a linear transformation to a state vector $T\tilde{\mathbf{x}} = \mathbf{x}$ does not change the input–output behavior of the system. The controllability Gramian W_C and observability Gramian W_O associated with the LTI system realized by (A, B, C) are defined as

$$W_C = \int_0^\infty e^{At} B B^* e^{A^* t} dt, \qquad (26)$$

and

$$W_0 = \int_0^\infty e^{A^*t} C^* C e^{At} dt,$$
 (27)

respectively. We look for the linear transformation on the state vector space that "balances" the system. In other words, we look for a projection matrix *T* such that

$$W'_{C} = T^{-1}W_{C}T^{-*} = \Sigma,$$

 $W'_{O} = T^{*}W_{O}T = \Sigma$ (28)

where $\Sigma = \operatorname{diag}(\sigma_1, \ldots, \sigma_M)$ with $\sigma_1 \ge \sigma_2 \ge \cdots \ge \sigma_M \ge 0$. The $\{\sigma_i\}$ are Hankel singular values of the dynamic system.

For a general matrix A and $C \neq B^T$, the transformation matrix T is computed as follows. First, Gramian matrices are computed by solving the Lyapunov equations $AW_C + W_CA^* = BB^*$ and

 $A^*W_0 + W_0A = C^*C$. Since the Gramian matrices W_C and W_0 are positive definite, the Cholesky decomposition of those matrices can be computed as $W_0 = L_0 L_0^*$ and $W_0 = L_0 L_0^*$ where L_C and L_0 are $M \times M$ lower triangular matrices. The Hankel singular values of the system are derived as singular values of $L_C^*L_0$, which are equal to the square root of the eigenvalues of W_0W_C and $L_C^*W_0L_C$. Consider the eigenvalue decomposition

$$L_C^* W_0 L_C = U^* \Sigma^2 U. (29)$$

Then, the transformation matrix is constructed as

$$T = L_c^* U^* \Sigma^{-\frac{1}{2}}, \quad T^{-1} = \Sigma^{\frac{1}{2}} U L_c.$$
 (30)

In the present case of a system with F(s) in (20) as the transfer function, matrix elements of W_C and W_0 can be analytically ob-

$$[W_C]_{ij} = \int_0^\infty e^{-a_i t} \sqrt{c_i c_j^*} e^{-a_j^* t} dt = \frac{\sqrt{c_i c_j^*}}{a_i + a_j^*},$$

$$[W_O]_{ij} = \int_0^\infty e^{-a_i t} \sqrt{c_i^* c_j} e^{-a_j t} dt = \frac{\sqrt{c_i^* c_j}}{a_i^* + a_j}$$

$$= [W_C]_{ij}^* = [W_C]_{ji}.$$
(31)

Thus, W_C is a self-adjoint quasi-Cauchy matrix, and $W_O = \overline{W}_C =$ W_C^T , where the overline indicates the element-wise complex conjugate of the matrix. The W_C can be decomposed as

$$W_C = \overline{U}\Sigma U^T, \tag{32}$$

where $U_M \in \mathbb{C}^{M \times M}$ with $U^T U = I$, and $\Sigma = \text{diag}(\sigma_1, \dots, \sigma_M)$ with $\sigma_i \geq \sigma_2 \geq \cdots \geq \sigma_M > 0$. A decomposition of the form (32) is referred to as a con-eigenvalue decomposition. Each column of *U* is called a con-eigenvector, and σ_i is called a con-eigenvalue. The coneigenvalues $\{\sigma_i\}$ (i = 1, ..., M) coincide with the Hankel singular values of the LTI system with F(s) as the transfer function, which can be verified as

$$W_0 W_C = W_C^T W_C = U \Sigma U^* \overline{U} \Sigma U^T = U \Sigma^2 U^{-1}. \tag{33}$$

A fast and accurate con-eigenvalue decomposition algorithm for a quasi-Cauchy matrix was developed by Haut and Beylkin [18]. In this algorithm, a rank-revealing Cholesky decomposition, $W_C =$ (PL)D²(PL)*, is computed using the highly accurate Gaussian elimination with a complete pivoting (GECP) algorithm for a quasi-Cauchy matrix in [26]. The con-eigenvalue decomposition is computed using the Demmel's SVD algorithm for a matrix with rankrevealing decomposition (RRD) [26,27]. This method allows us to compute con-eigenvalues and con-eigenvectors with high relative accuracy.

In the mxpfit library, a modified balanced truncation algorithm combined with the fast con-eigenvalue solver described above was implemented, as shown in Algorithm 3. The truncation is made in step 2, where the order of truncated system, M', is determined by the error bound computed from the Hankel singular values. Then, the first linear transformation is performed in step 3. The second transformation is made in steps 4 and 5 so that the transfer function of the reduced system has the form (21).

Algorithm 3 (Modified Balanced Truncation Method). Input: Sequences $\{a_i\}$ (Re $(a_i) > 0$) and $\{c_i\}$ for i = 1, ..., M to define transfer function F(s) in (20), and a prescribed accuracy $\epsilon > 0$. Output: Sequences $\{\hat{a}_i\}$ and $\{\hat{c}_i\}$ for $i=1,\ldots,M'(M'\leq M)$ defining the transfer function of reduced system $\hat{F}(s)$ in (21).

1. Compute the con-eigenvalue decomposition of a quasi-Cauchy matrix $W_C = \overline{U}\Sigma U^T$ using the fast and accurate algorithm in [18].

- Find M' such that 2∑_{i=M'+1}^M σ_i ≤ ε.
 Compute an M' × M' matrix A' = U_{M'}*AŪ_{M'} and a vector of length M', B' = U_{M'}*B, where U_{M'} = U[1 : M, 1 : M'].
- 4. Compute the eigenvalue decomposition of $A' = X \Lambda X^{-1}$, and set $\hat{a}_i = \Lambda_{ii}$ for $i = 1, \dots, M'$. Since matrix A' is symmetric, eigenvectors can be chosen such that $X^{-1} = X^{T}$.
- 5. Compute $B'' = X^{-1}B' = X^TB'$ and set $\hat{c}_i = (B_i'')^2$ for i =

4. Description of the program

The mxpfit library is a pure template and header-only library written in C++. The library depends on the Eigen linear algebra library (version 3.3 or later) [28], and FFTW3 library for fast Fourier transforms [29,30]. The reason for using the Eigen library as the backend is that it provides a generic application programming interface (API) for matrix vector arithmetic and matrix decompositions using C++ templates. This feature enables us to conduct a simulation using various scalar types including single/double/quadruple precision floating-point numbers and custom scalar types such as multiprecision numbers implemented in the GNU Multi-Precision Library (GMP).

One can use this library by including just header files. To compile the source code using mxpfit, a modern C++ compiler supporting the C++11 standard is necessary. The FFTW3 library with the appropriate scalar type has to be linked using the fast ESPRIT algorithm. For building an example and test codes, CMake and related build tools such as GNU make are also required.

In this library, an exponential sum function of the form (1) is represented by ExponentialSum class in which arrays to store parameters a_i and c_i are dynamically allocated. Scalar types for an exponent a_i and weight c_i should be provided as template arguments (the scalar type for a weight is same as that of an exponent by default). Alternatively, one can view two separated arrays as an exponential function via ExponentialSumWrapper class. A utility function to make an instance of this class easily is provided. An instance of ExponentialSumWrapper cannot be resized or modified. Users are responsible to manage the memory allocation and to set valid values of input arrays. Both the ExponentialSum and Exponential SumWrapper classes have common interface to access those parameters and that to evaluate function value at a given coordinate (t) which is inherited from ExponentialSum-Base class. See the right-half part of Fig. 2 for relationship between these classes. These classes can be used for instance,

```
// Create a five-term exponential sum with real exponents
// and real weights.
mxpfit::ExponentialSum<double> expsum(5);
assert(expsum.size() == 5);
// You can change the number of terms dynamically.
expsum.resize(2);
assert(expsum.size() == 2);
// Access to parameters
expsum.exponent(0) = 1.0; expsum.weight(0) = 0.5;
expsum.exponent(1) = 2.0; expsum.weight(1) = 1.5;
// Evaluate function value
auto f = expsum(2.5);
// Create a view of a complex exponential sum from pre-existing
// arrays of exponents and weights
using Complex = std::complex<double>;
Eigen::ArrayXcd a(2), c(2);
a << Complex(1.0, 0.0), Complex(2.0, 0.0); // exponents
c << Complex(0.5, 0.0), Complex(1.5, 0.0); // weights</pre>
// This will return an instance of
// mxpfit::ExponentialSumWrapper.
auto expsum_wrapper = mxpfit::makeExponentialSum(a, c);
assert(expsm_wrapper.size() == 2);
```

```
// Evaluate function value
auto f = expsum(2.5);

// You cannot modify ExponentialSumWrapper directly.
// Modify the arrays (a,c) and re-create an instance.
// expsum_wrapper.resize(3);  /* not allowed */
// expsum_wrapper.exponent(0) = 2.0; /* not allowed */
```

The library provides two main APIs: FastESPRIT class for finding the best parameters in (1) from sampled data on a uniform grid via the modified fast ESPRIT algorithm, and BalancedTruncation class for finding optimal parameters from a given exponential sum function via the modified balanced truncation method. Both classes return an instance of ExponentialSum class. The FastESPRIT class internally uses the partial Lanczos bidiagonalization algorithm and the CGLS method with Vandermonde preconditioner as has been described in 2, which have been implemented in separate classes. Precisely speaking, the FastESPRIT class depends on the classes described below. The relationships between those classes are visualized in Fig. 1 using a unified modeling language (UML) class diagram.

- PartialLanczosBidiagonalization: computes a lowrank approximation of a given matrix via a partial Lanczos bidiagonalization algorithm.
- HankelMatrix: represents a rectangular Hankel matrix (4), with an interface for the Hankel matrix-vector product described in Algorithm 2. This class internally uses the FFTW3 library. MatrixFreeGEMV < HankelMatrix > is provided as an argument of PartialLanczosBidiagonalization.
- VandermondeLeastSquareSolver solves an overdetermined Vandermonde system (14) using the CGLS method.
 This is a realization of the CGLS solver provided by Eigen, where MatrixFreeGEMV < VandermondeMatrix and VandermondePreconditioner are used as the matrix type and preconditioner type, respectively.</p>
- VandermondePreconditioner: solves a preconditioning linear system for a Vandermonde matrix (18).
- VandermondeMatrix: represents a rectangular Vandermonde matrix (15), with an interface for the Vandermonde matrix-vector product.
- MatrixFreeGEMV: provides interface overloading operator* for the product of the HankelMatrix/Vandermonde-Matrix and any vector type in Eigen.

A typical usage of the FastESPRIT class is as follows:

```
using Complex = std::complex<double>
using ESPRIT = mxpfit::FastESPRIT<double>;
// 1. Preparation:
   N: number of grid points (N >= 1)
//
     L: window length (1 <= L <= N)
//
     M_max: upper bound of M, the number of terms
            (1 \le M_{max} \le min(N, N - L + 1))
//
ESPRIT esprit(N, L, M_max);
// 2. Fitting:
//
     f: sample data on uniform grid
//
         (Eigen's vector of length N)
//
     t0: first grid point
     h: interval between two grid points (h > 0)
     epsilon: prescribed accuracy (0 < epsilon << 1)
// * Result type is mxpfit::ExponentialSum<Complex>
auto expsum = esprit.compute(f, t0, h, epsilon);
// You can re-use 'esprit' for fitting other data with
// same length
auto expsum2 = esprit.compute(f2, t0, h, epsilon);
```

As the first step, one needs to create an object of the FastE-SPRIT class with a number of grid points N, window length L, and upper bound of M for exponential sum (1). In this step, memory space for internal arrays is allocated, and optimal "plans" for an FFT and inverse FFT of length N are generated. Then, parameters of the exponential sum approximation are computed by calling the compute method of this object with sample data \boldsymbol{f} , parameters t_0 and h which define a uniform grid $t_k = t_0 + hk$ ($k = 0, \ldots, N-1$), and a prescribed accuracy ϵ as input arguments. Any vector expressions available in the Eigen library can be used as an argument type of \boldsymbol{f} .

In the BalancedTruncation class, a rank-revealing decomposition $W_C = (PL)D^2(PL)^*$ of a self-adjoint quasi-Cauchy matrix (31) is computed, then coneigenvalue decomposition (32) is performed. These algorithms are also implemented in separate classes described as follows. The relationships between those classes are shown in Fig. 2 as a UML class diagram.

- SelfAdjointQuasiCauchyRRD: computes a rankrevealing Cholesky decomposition of a self-adjoint quasi-Cauchy matrix with high relative accuracy.
- SelfAdjointConeigensolver: computes the coneigenvalue decomposition of a self-adjoint matrix having an RRD, A = XD²X*, with high relative accuracy.

The BalancedTruncation class can be used as follows:

```
using Complex = std::complex<double>;
mxpfit::ExponentialSum<Complex> f_orig(N);
// ... set 'f_orig' here ...
mxpfit::BalancedTruncation<Complex> truncation;
// This returns a 'ExponentialSum' object with same scalar type
// epsilon: prescribed accuracy (0 < epsilon << 1)
auto f_truncated = truncation.compute(f_orig, epsilon);</pre>
```

The BalancedTruncation class can be applied to an exponential sum with real parameters as long as $a_j>0$ and $c_j>0$ are satisfied. Note that in case any $c_j<0$, the truncation fails because a domain error occurs in the built-in square-root function (std::sqrt).

5. Numerical examples

In this section, numerical examples for the fast ESPRIT method and balanced truncation method are provided. The calculations were made on a laptop computer with an Intel Core i7 processor @ 3.3 GHz. Programs were built using clang++ compiler with '-std=c++11 -stdlib=libc++ -03 -march=native - DNDEBUG' flags.

5.1. Example 1 (Fast ESPRIT applied to analytical function)

We consider a nonlinear approximation of the function on a finite interval,

$$g(t) = \operatorname{sinc}(t) = \frac{\sin(t)}{t} \quad (t \in [0, t_{\max}])$$
 (34)

by an exponential sum. The fast ESPRIT algorithm was applied with varying numbers of sample data N from 2^8 to 2^{18} for sampled values $g(t_k)$ with $t_k = hk$, h = 1/16 (k = 0, ..., N-1). Other arguments were chosen as L = N/2, $M_{\text{max}} = \min(L, 500)$, and $\epsilon = 10^{-12}$. The exponential sum of the form (1) was obtained with complex exponents a_i and coefficients c_i .

complex exponents a_j and coefficients c_j . In the case of $N=2^{12}=4096$, we obtained the exponential sum $\tilde{g}(t)=\sum_{j=1}^M c_j e^{-a_i t}$ for a sinc function of order M=30 on the interval $t\in[0,256]$. The function f(t) and absolute error $|g(t)-\tilde{g}(t)|$ on this interval are shown in Fig. 3. The absolute error is kept smaller than the prescribed accuracy 10^{-12} .

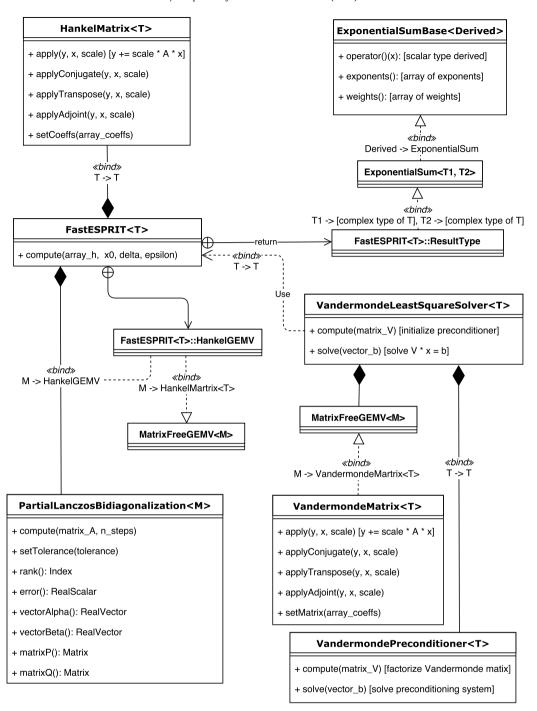


Fig. 1. Fast ESPRIT class diagram.

The order of exponential sum, approximation errors and the running time of the fast ESPRIT algorithm for a sinc function with different N are summarized in Table 1. The maximum absolute error is defined as $\max|g(t_k)-\tilde{g}(t_k)|$. The running time for preparing a FastESPRIT object and that for the fitting data are listed separately. As can be seen, most of the CPU time is spent on the preparation step. In fact, most of the CPU time is spent on creating plans for the FFTW3 library. Once a FastESPRIT object is prepared, a nonlinear fitting can be obtained efficiently. Even in the case of $N=2^{18}=262$ 144, it only takes about 2 s to obtain the fitting data. The present implementation of the fast ESPRIT algorithm is feasible for fitting many numbers of data sets with the same length N.

5.2. Example 2 (Fast ESPRIT applied to numerically sampled data)

Let us consider the exponential sum of order M = 5,

$$h(t) = \sum_{j=1}^{5} c_j e^{-if_j t}$$
 (35)

with

$$\mathbf{f} = (0, \pi/4, -\pi/4, \pi/2, -\pi/2)^{\mathrm{T}}, \tag{36}$$

$$\mathbf{c} = (34, 300, 300, 1, 1)^{T}, \tag{37}$$

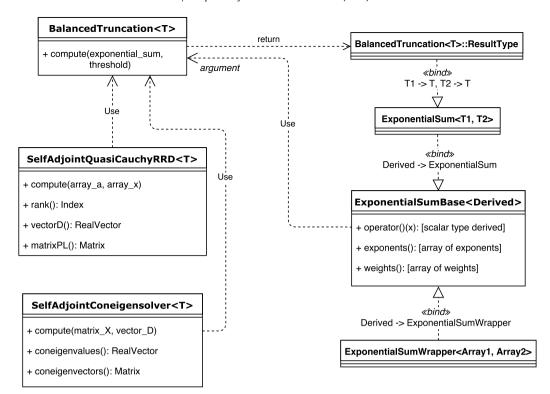


Fig. 2. Balanced truncation class diagram.

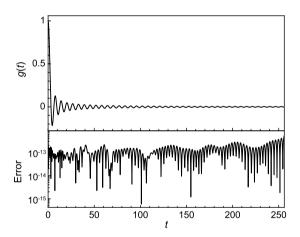


Fig. 3. Function $g(t) = \operatorname{sinc}(t)$ (upper) and absolute error between g(t) and exponential sum approximation $\tilde{g}(t)$ of order 30 on the interval [0, 256].

Table 1 Errors and running time of fast ESPRIT algorithm for $g(t_k) = \text{sinc}(t_k)$ with $t_k = k/16$, (k = 0, ..., N-1). Running time during preparation step and fitting step are listed separately (in milliseconds).

N	М	Max abs. error	Time (prep.)	Time (fit)
2 ⁸	12	1.1×10^{-11}	38	0.3
2^{9}	16	2.4×10^{-12}	50	0.4
2^{10}	22	2.1×10^{-13}	79	1
2^{11}	26	3.0×10^{-13}	131	3
2^{12}	30	3.5×10^{-13}	248	6
2^{13}	34	4.8×10^{-13}	498	15
2^{14}	38	2.1×10^{-13}	966	43
2^{15}	42	1.7×10^{-13}	1,841	163
2^{16}	46	3.4×10^{-13}	3,747	421
2^{17}	49	4.2×10^{-13}	7,923	792
218	53	2.9×10^{-13}	17,742	1,958

and construct sampling data on uniform grid as

$$h_k = h(k) + e_k = 34 + 600 \cos\left(\frac{\pi t}{4}\right) + 2\cos\left(\frac{\pi t}{2}\right) + e_k \quad (k = 0, 1, ..., N - 1),$$
(38)

where e_k are random noise uniformly distributed in [-3, 3]. The same sequence has been investigated in Example 6.1 in [6]. It should be noted that the noise is higher than the amplitude of coefficient related to the frequencies $\pm \pi/2$. The frequencies \tilde{f} and coefficients \tilde{c} were estimated from the sequence $\{h_k\}$ by applying modified fast ESPRIT algorithm with the number of sampling points, N, varying from 2^8 to 2^{18} . The window size was chosen L =N/2 and target accuracy was set to $\epsilon = 10^{-5}$. The maximum error of frequencies, max $(|f_1 - \tilde{f}_1|, ..., |f_5 - \tilde{f}_5|)$, as well as the maximum errors of coefficients, max $(|c_1 - \tilde{c}_1|, ..., |c_5 - \tilde{c}_5|)$, where the maximum are formed over 10 trials. The same measurements were also performed by original (sub-optimal) ESPRIT algorithm with N up to 2^{13} . The results are summarized in Table 2. The maximum errors observed in the results of fast ESPRIT algorithm and original one are almost same. The maximum error of frequencies occurs at $f_4 = \pi/2$ and $f_5 = -\pi/2$. The errors of other frequencies are smaller about two order of magnitude. Both the errors of frequencies and coefficients decrease with increase of N.

The CPU time of fast ESPRIT algorithm and original ESPRIT algorithm averaged over 10 trials are also shown in Table 2, in which the running time for preparation step of fast ESPRIT algorithm is excluded. We estimated computational complexities of these two algorithms by performing curve fittings. The CPU time of original ESPRIT algorithm is found to be proportional to N^b with b=2.93, as has been expected from the complexity of the dense SVD. The CPU time of fast ESPRIT algorithm is much smaller than that of original ESPRIT algorithm. The complexity of fast ESPRIT algorithm is proportional to $N \log N$. The result suggests that most of CPU time is spent for the Hankel matrix–vector multiplications in the partial Lanczos bidiagonalization step.

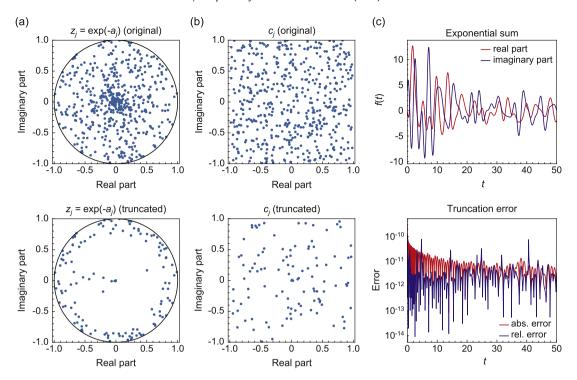


Fig. 4. Results of the modified balanced truncation applied to a randomly generated exponential sum of order 500. (a) Distribution of $z_j = e^{-a_j}$ for original and truncated exponential sum. (b) Distribution of coefficients c_j for original and truncated exponential sum. (c) The function f(t), the absolute truncation error $|f(t) - \hat{f}(t)|$, and relative truncation error $|f(t) - \hat{f}(t)|/|f(t)|$ are on the interval $t \in [0, 50]$. (For interpretation of the references to color in this figure legend, the reader is referred to the web version of this article.)

Table 2 Errors and running time of fast ESPRIT algorithm for Example 5.2 with different number of sampling points, N, which are compared with the values obtained by the original ESPRIT algorithm using full SVD. The window size is chosen to be L = N/2, and accuracy is set to $\epsilon = 10^{-5}$.

N	Max error of free	Max error of frequencies		Max error of coefficients		Time (ms)	
	Fast	Original	Fast	Original	Fast	Original	
28	2.81×10^{-3}	2.95×10^{-3}	4.40×10^{-1}	4.21×10^{-1}	0.1	5.8	
2^{9}	9.10×10^{-4}	9.22×10^{-4}	3.61×10^{-1}	3.52×10^{-1}	0.3	18.8	
2^{10}	2.75×10^{-4}	3.08×10^{-4}	1.93×10^{-1}	2.31×10^{-1}	0.4	87.6	
2^{11}	1.07×10^{-4}	1.08×10^{-4}	1.76×10^{-1}	1.75×10^{-1}	0.7	561.3	
2^{12}	5.72×10^{-5}	4.03×10^{-5}	1.34×10^{-1}	1.35×10^{-1}	1.3	4748.9	
2^{13}	1.78×10^{-5}	1.76×10^{-5}	7.64×10^{-2}	7.63×10^{-2}	2.5	36003.7	
2^{14}	4.98×10^{-6}	_	4.42×10^{-2}	_	5.9	_	
2^{15}	1.85×10^{-6}	_	4.99×10^{-2}	_	15.8	_	
2^{16}	7.12×10^{-7}	_	2.33×10^{-2}	_	32.1	_	
2^{17}	1.99×10^{-7}	_	1.96×10^{-2}	_	67.7	_	
2^{18}	6.44×10^{-8}	_	1.60×10^{-2}	_	144.3	_	

5.3. Example 3 (modified balanced truncation)

The modified balanced truncation algorithm was applied to a randomly generated exponential sum in (1) of order M=500. The complex exponents were chosen such that $z_j=e^{-a_j}=r_je^{i\delta_j}$ with uniformly distributed $r_j\in[0,1]$, and $\delta_j\in[0,2\pi]$. All z_j were inside the unit disk on a complex plane. The complex coefficients c_j were uniformly distributed on [-1,1]+i[-1,1]. The truncation was made with the prescribed accuracy $\epsilon=10^{-12}$.

Fig. 4 summarizes the results of a typical run. A truncated exponential sum $\hat{f}(t)$ of order M'=133 was obtained. Parameters $z_j=e^{-a_j}$ and c_j before and after the truncation are shown in Fig. 4(a) and (b). The function f(t) is shown in the upper panel of Fig. 4(c). The real and imaginary parts of f(t) are indicated by red and blue lines, respectively. The absolute truncation error $|f(t)-\hat{f}(t)|$ and relative truncation error $(|f(t)-\hat{f}(t)|)/|f(t)|$ are given at the bottom panel of Fig. 4(c). The relative error is kept below 10^{-11} at most points, which is slightly higher than the prescribed accuracy ϵ . The CPU time spent on the truncation was 2.1 s.

Acknowledgments

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