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# SYSTEMATIC ERRORS OF THE LIDFT METHOD: ANALYTICAL FORM AND VERIFICATION BY A MONTE CARLO METHOD

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#### Abstract

This paper derives analytical formulas for the systematic errors of the linear interpolated DFT (LIDFT) method when used to estimating multifrequency signal parameters and verifies this analysis using Monte-Carlo simulations. The analysis is performed on the version of the LIDFT method based on optimal approximation of the unit circle by a polygon using a pair of windows. The analytical formulas derived here take the systematic errors in the estimation of amplitude and frequency of component oscillations in the multifrequency signal as the sum of basic errors and the errors caused by each of the component oscillations. Additional formulas are also included to analyze particular quantities such as a signal consisting of two complex oscillations, and the analyses are verified using Monte-Carlo simulations.

Keywords: LIDFT, multifrequency signal, interpolated DFT, spectrum estimation, zero padding, unit circle, approximation by polygon.

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# **1. Introduction**

Modern measurement techniques use sophisticated methods based on digital signal processing (DSP) that have become increasingly sophisticated to handle more complex mathematical problems [1-6]. An important class of complex problems is spectrum estimation, understood as the estimation of multifrequency signal parameters. The most accurate methods of this type of estimation are the Prony method and its modifications, the transmittance modeling methods and subspace methods (based mainly on the properties of the signal autocorrelation matrix and its eigenvalues) [1-4, 7-8]. The interpolated spectrum methods yield less accurate results but require a significantly shorter computation time. This class of methods includes nonparametric spectrum interpolation methods [9-16] and interpolated methods based on DFT [17-44]. A special class is non-iterative methods that take the leakage of the spectrum into account in their equations, including Multipoint Weighted Interpolated DFT (MWIDFT) methods [32, 34, 37] and the Linear Interpolated DFT (LIDFT) method [29-31, 41-42, 45-46]. The group of MWIDFT methods is only defined for the maximum sidelobe decay windows (class I Rife-Vincent windows) [17, 34, 37, 39]. The most advanced MWIDFT method appears to be the one based on complex spectrum values that approximate the spectrum leakage with a polynomial and a step 1 bin [39]. However, the MWIDFT methods have limitations, mainly with respect to frequency resolution and the level of oscillation amplitude versus the level of spectrum leakage. The LIDFT method can be used where former interpolation methods have failed by considering the spectrum leakage. The LIDFT method has evolved, and the latest version [42] is based on optimal approximation of the unit circle by a polygon [42, 45-46] and the use of a pair of windows [42]. The current

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# paper addresses the systematic errors of this latest version of the LIDFT method. 2. Unit circle approximation and the pair of windows in the LIDFT method

Let us assume that the aim of estimation is to determine the parameters  $A_k$ ,  $\omega_k = 2\pi f_k$ , and  $\varphi_k$  of the following multifrequency signal, defined in the domain of time *t*:

$$y(t) = \sum_{k \in S} A_k \sin(\omega_k t + \varphi_k), \quad S = \{k_1, ..., k_K\},$$
(1)

where the assumption  $k = k_1, ..., k_K$  (in which the condition  $k_{i+1} = k_i+1$  is not required) instead of k = 1, ..., K allows for the integer index k to be linked with the appropriate integer index of the DFT spectrum sample. This assumption involves a simple renumbering of components that simplifies the description of the LIDFT method by coding additional information about the rough location of components in the DFT spectrum within the index k.

Equation (1) can be written by:

$$y(t) = \sum_{k \in S_1} B_k e^{j\omega_k t} , \quad S_1 = \{k_1, \dots, k_P\},$$
(2)

where:  $B_k = (A_k/(2j))e^{j\varphi_k}$ , P = 2K,  $B_{k_{P-m+1}} = B_{k_m}^*$ ,  $e^{j\varphi_{k_{P-m+1}}} = (e^{j\varphi_{k_m}})^*$ , m = 1, ..., K.

Signal (2) is sampled with a frequency  $f_s = 1/T$  ( $f_s/2 > \max_k \{f_k\}$ , where  $f_k = \omega_k/(2\pi)$ ) results in *N* samples (indexed in general by  $n = n_0, ..., n_0 + N - 1$  from the starting point  $n_0$ ) and has the form:

$$y_n = y(t = nT) = \sum_{k \in S_1} B_k e^{j2\pi f_k nT} = \sum_{k \in S_1} B_k e^{j2\pi n\lambda_k/N}, \quad S_1 = \{k_1, \dots, k_P\},$$
(3)

where  $\lambda_k = Nf_k/f_s$  is the normalized frequency in the DFT bin and  $B_k$  is the complex amplitude of the *k*-th complex oscillation  $B_k e^{j2\pi n\lambda_k/N}$ . The starting point  $n_0$  for the sample index *n* theoretically does not change the frequency and amplitude estimation results when the spectrum leakage is negligible but only changes the phase (which can be easy recalculated for different values of  $n_0$ ). Most often  $n_0 = 0$ , but sometimes it is useful to assume that  $n_0 = -N/2$ when *N* is even to make the time axis symmetrical, simplifying many equations and algorithm derivations.

The LIDFT method presented in [42] makes two assumptions:

1) Approximation of the unit circle by a polygon (with the parameter  $R \ge 1$  that allows the number of sides of the approximation polygon to be varied) [45, 50, 51], defined by:

$$e^{-j2\pi n\lambda_k/N} \approx e^{-j2\pi nk/M} [\alpha_n + j\gamma_k \beta_n], \quad M = NR, \qquad (4)$$

$$\lambda_{k} = \frac{1}{R} (k + \gamma_{k}), \quad \gamma_{k} \in [-1/2, 1/2],$$
(5)

where  $k \in S_1$  (here, the indexing used in (1) is very useful) and the parameter *R* is selected such that *M* is an integer (usually as a power of 2 when the *radix*-2 FFT algorithm is used in the final estimation method).  $\alpha_n$  and  $\beta_n$  are defined with parameters  $\eta_1$  and  $\eta_2$ , respectively:

$$\alpha_n(\eta_1) = (1 - \eta_1) \cos x_n + \eta_1, \quad \beta_n(\eta_2) = -2 \cdot [(1 - \eta_2) \sin x_n + \eta_2 \tan x_n], \tag{6}$$

where:

$$n = -N/2, ..., N/2 - 1, \quad x_n = \pi n/M$$
 (7)

and optimal parameters values  $\eta_1$  and  $\eta_2$  that minimize maximum approximation errors should be found in the triangle on the plane  $(\eta_1, \eta_2)$  given by the following conditions [42, 45, 46]:

$$1/4 \le \eta_1 \le 1/2, \quad 1/12 \le \eta_2 \le 1/3, \quad \eta_2 - \eta_1 \le -1/6.$$
 (8)

2) The pair of windows, the even window  $h_n$  and the odd window  $g_n$ , are obtained from the prototype window  $w_n$  by:

$$h_n = w_n^2 \alpha_n, \quad g_n = w_n^2 \beta_n. \tag{9}$$

Simulations performed with  $w_n$  as a triangular window [42, 45, 46] show that parameter values  $\eta_1 \approx 1/2$  and  $\eta_2 \approx 1/6$  are close to optimal for this window, but when other windows are used, these values should be verified as being close to optimal.

After applying the above two assumptions (unit circle approximation and the pair of windows) in DFT formulas and denoting  $C_k = \gamma_k B_k$  with  $\gamma_k = R\lambda_k - k$  obtained from (5), the linear matrix equation is derived [42], which has the form  $\mathbf{Fx} = \mathbf{f}$ :

$$\begin{bmatrix} r_{i-k} & s_{i-k} \\ p_{i-k} & q_{i-k} \end{bmatrix}_{2P \times 2P} \begin{bmatrix} B_k \\ C_k \end{bmatrix}_{2P \times 1} = \begin{bmatrix} u_i \\ v_i \end{bmatrix}_{2P \times 1}, \quad i,k \in S_1,$$
(10)

where  $S_1$  is the set of initial locations of components in the spectrum, *i.e.*, (5) implies the condition for each integer  $k \in S_1$ :

$$\lambda_k \in [(k - 0.5) / R, (k + 0.5) / R].$$
(11)

The elements of the matrices in (10) are given by:

$$r_m = \operatorname{FFT}_m \{ \alpha_n h_n \}_{NR}, \quad s_m = -j \operatorname{FFT}_m \{ \beta_n h_n \}_{NR}, \quad p_m = j \operatorname{FFT}_m \{ \alpha_n g_n \}_{NR}, \quad (12)$$

$$q_{m} = FFT_{m} \{\beta_{n}g_{n}\}_{NR}, \quad u_{m} = FFT_{m} \{y_{n}h_{n}\}_{NR}, \quad v_{m} = jFFT_{m} \{y_{n}g_{n}\}_{NR}, \quad (13)$$

$$m = 0, \dots, NR - 1,$$
 (14)

where  $FFT_m\{z_n\}_{NR}$  denotes the *m*-th element of the set of values obtained by the FFT algorithm applied to the set of *N* values  $z_n$  supplemented with zeros to form the *NR* element set as follows:

$$\{z_n\}_{NR} = \{z_0, \dots, z_{N/2-1}, 0, \dots, 0, z_{-N/2}, \dots, z_{-1}\}_{NR}.$$
(15)

The solution of (10) allows for the direct determination of the amplitudes  $B_k$  and the normalized frequencies  $\lambda_k$  from:

$$\lambda_k = \frac{k + \operatorname{Re} \gamma_k}{R}, \quad \gamma_k = C_k / B_k, \quad (16)$$

where  $\text{Re}_{\gamma_k}$  is introduced instead of  $\gamma_k$  because the result is usually not real but is complex due to the presence of noise in the signal, the finite accuracy of calculations and the possible nonideal initial location of the components of the spectrum. When this initial location is properly defined, then  $|\text{Re}_{\gamma_k}| \le 1/2$ , and if it is not properly defined, several iterations can be performed [41]. The values  $\text{Im}_{\gamma_k}$ , which should be close to 0, can also be used to verify that the initial locations of the components are correct.

#### 3. Systematic errors of the LIDFT method for the case of a multifrequency signal

For the signal consisting of *P* complex oscillations (each with a complex amplitude  $B_k$  and a normalized frequency  $\lambda_k$ ) and with condition (11), the signal samples  $y_n$  are defined by (3), and applying (10) and (12)-(14) allows us to obtain the estimators  $\hat{B}_k$  and  $\hat{C}_k$  instead of the exact values  $B_k$  and  $C_k = B_k \gamma_k$  due to the systematic errors of the LIDFT method. (10), taking (3) into account, takes on the form:

$$\begin{bmatrix} r_{i-k} & s_{i-k} \\ p_{i-k} & q_{i-k} \end{bmatrix}_{2P\times 2P} \begin{bmatrix} \hat{B}_k \\ \hat{C}_k \end{bmatrix}_{2P\times 1} = \begin{bmatrix} H_{i-k}(-\gamma_k) \\ G_{i-k}(-\gamma_k) \end{bmatrix}_{2P\times P} \begin{bmatrix} B_k \end{bmatrix}_{P\times 1}, \quad i,k \in S_1,$$
(17)

in which  $H_{i-k}(-\gamma_k)$  and  $G_{i-k}(-\gamma_k)$  are defined by:

$$H_m(\gamma) = H\left(\frac{m+\gamma}{R}\right), \quad G_m(\gamma) = G\left(\frac{m+\gamma}{R}\right), \quad |\gamma| \le 1/2,$$
(18)

where m = i-k is the integer number,  $\gamma = -\gamma_k$ , functions  $H(\lambda)$  and  $G(\lambda)$  are defined for any real  $\lambda$ , are periodic with period N (*e.g.*,  $H(\lambda+N) = H(\lambda)$ ) and are shifted Discrete-time Fourier Transforms (DtFT) of N samples of windows  $h_n$  and  $jg_n$ :

$$H(\lambda) = \sum_{n=-N/2}^{N/2-1} h_n e^{-j2\pi n\lambda/N} , \quad G(\lambda) = j \sum_{n=-N/2}^{N/2-1} g_n e^{-j2\pi n\lambda/N} .$$
(19)

By introducing  $M_{mk}$  and  $N_{mk}$  according to:

$$\begin{bmatrix} r_{i-m} & s_{i-m} \\ p_{i-m} & q_{i-m} \end{bmatrix}_{2P \times 2P} \begin{bmatrix} M_{mk} \\ N_{mk} \end{bmatrix}_{2P \times P} = \begin{bmatrix} H_{i-k}(-\gamma_k) \\ G_{i-k}(-\gamma_k) \end{bmatrix}_{2P \times P}, \quad i, m, k \in S_1,$$
(20)

based on (17) and (20):

$$\hat{B}_k = \sum_{i \in S_1} M_{ki} B_i , \qquad \hat{C}_k = \sum_{i \in S_1} N_{ki} B_i .$$
 (21)

For  $\hat{B}_k$ , the maximum (for the worst-case phase arg  $B_i$ ) of the systematic error  $\delta_d |B_k|$  of  $|B_k|$  is derived from (21) (App. A):

$$\delta_{\mathrm{d}} \mid B_k \mid = \delta_{\mathrm{d}k} \mid B_k \mid + \sum_{i \in S_1, i \neq k} \delta_{\mathrm{d}i} \mid B_k \mid,$$
(22)

where:

$$\delta_{dk} | B_k | = | M_{kk} - 1 |, \qquad (23)$$

$$\delta_{di} | B_k | = \frac{|B_i|}{|B_k|} \cdot \delta'_{di} | B_k |, \qquad \delta'_{di} | B_k | = |M_{ki}|$$
(24)

are the components of total error (22).

Analogously to (22)-(24), the maximum (for the worst-case phase arg  $B_i$ ) of the systematic error  $\Delta_d \lambda_k$  of  $\lambda_k$  is derived (App. B):

$$\Delta_{d}\lambda_{k} = \Delta_{dk}\lambda_{k} + \sum_{i\in S_{1}, i\neq k}\Delta_{di}\lambda_{k} , \qquad (25)$$

where:

$$\Delta_{dk}\lambda_{k} = R^{-1} | (N_{kk} - \gamma_{k}) - \gamma_{k}(M_{kk} - 1) |, \qquad (26)$$

$$\Delta_{di}\lambda_{k} = \frac{|B_{i}|}{|B_{k}|} \cdot \Delta_{di}^{\prime}\lambda_{k}, \qquad \Delta_{di}^{\prime}\lambda_{k} = R^{-1}|N_{ki} - \gamma_{k}M_{ki}|$$
(27)

are the components of total error (25).

To determine (22)-(27),  $N_{mk}$  and  $M_{mk}$  must be known from (20). The right side of this equation, taking (12)-(13) and (18)-(19) into account, has the form:

$$\begin{bmatrix} H_{i-k}(-\gamma_{k}) \\ G_{i-k}(-\gamma_{k}) \end{bmatrix}_{2P\times P} = \begin{bmatrix} r_{i-k} & s_{i-k} \\ p_{i-k} & q_{i-k} \end{bmatrix}_{2P\times 2P} \begin{bmatrix} \mathbf{I} \\ \mathbf{\Lambda}_{\gamma} \end{bmatrix}_{2P\times P} - \begin{bmatrix} \Delta r_{i-k}(\gamma_{k}) & \Delta s_{i-k}(\gamma_{k}) \\ \Delta p_{i-k}(\gamma_{k}) & \Delta q_{i-k}(\gamma_{k}) \end{bmatrix}_{2P\times 2P} \begin{bmatrix} \mathbf{I} \\ \mathbf{\Lambda}_{\gamma} \end{bmatrix}_{2P\times P},$$

$$i, k \in S_{1}, \qquad (28)$$

where I is an identity matrix,  $\Lambda_{\gamma} = [\gamma_k]$  is the diagonal matrix with elements  $\gamma_k$  ( $k \in S_1$ ), and using notation from (12)-(13) and m = i-k, the new symbols introduced in (28) are defined as:

$$\Delta r_m(\gamma_k) = \text{FFT}_m\{\Delta \alpha_n(\gamma_k)h_n\}_{NR}, \quad \Delta s_m(\gamma_k) = -j\text{FFT}_m\{\Delta \beta_n(\gamma_k)h_n\}_{NR}, \quad (29)$$

$$\Delta p_m(\gamma_k) = j \text{FFT}_m \{ \Delta \alpha_n(\gamma_k) g_n \}_{NR}, \quad \Delta q_m(\gamma_k) = \text{FFT}_m \{ \Delta \beta_n(\gamma_k) g_n \}_{NR}, \quad (30)$$

$$\Delta \alpha_n(\gamma_k) = \alpha_n - \cos(2\gamma_k x_n), \quad \Delta \beta_n(\gamma_k) = \beta_n + \gamma_k^{-1} \sin(2\gamma_k x_n). \tag{31}$$

Applying (28) to (20) yields the following result:

$$\begin{bmatrix} r_{i-m} & s_{i-m} \\ p_{i-m} & q_{i-m} \end{bmatrix}_{2P \times 2P} \begin{bmatrix} [M_{mk}] - \mathbf{I} \\ [N_{mk}] - \mathbf{\Lambda}_{\gamma} \end{bmatrix}_{2P \times P} = -\begin{bmatrix} \Delta r_{i-k}(\gamma_k) & \Delta s_{i-k}(\gamma_k) \\ \Delta p_{i-k}(\gamma_k) & \Delta q_{i-k}(\gamma_k) \end{bmatrix}_{2P \times 2P} \begin{bmatrix} \mathbf{I} \\ \mathbf{\Lambda}_{\gamma} \end{bmatrix}_{2P \times P},$$
  
$$i, m, k \in S_1.$$
(32)

The result from (32), together with those from (22)-(27), form the basis for a more detailed analysis of the systematic errors of the LIDFT method for the case of two complex oscillations, presented in Section 4.

### 4. The case of two complex oscillations

Consider the case of two complex oscillations:  $B_k e^{j2\pi n\lambda_k/N}$ ,  $B_i e^{j2\pi n\lambda_i/N}$ , where  $\lambda_i > \lambda_k$ ,  $\lambda_k = (k+\gamma_k)/R$ ,  $\lambda_i = (i+\gamma_i)/R$ ,  $|\gamma_i| \le 1/2$ ,  $|\gamma_k| \le 1/2$ , the numbers *i*, *k*,  $\tau = i-k$  are integer values and  $S_1 = \{k, i\}$ . For this case, (22)-(27) have the form:

$$\begin{bmatrix} \delta_{d} \mid B_{k} \mid \\ \Delta_{d} \lambda_{k} \end{bmatrix} = \mathbf{D} \cdot \begin{bmatrix} 1 \\ \mid B_{i} \mid / \mid B_{k} \mid \end{bmatrix}, \quad \mathbf{D} = \begin{bmatrix} \delta_{dk} \mid B_{k} \mid & \delta'_{di} \mid B_{k} \mid \\ \Delta_{dk} \lambda_{k} & \Delta'_{di} \lambda_{k} \end{bmatrix},$$
(33)

$$\mathbf{D}(\tau, \gamma_k, \gamma_i) = \operatorname{abs}\left(\begin{bmatrix} 1 & 0 \\ -\gamma_k / R & 1/R \end{bmatrix} \mathbf{M}\right), \quad \mathbf{M} = \begin{bmatrix} (M_{kk} - 1) & M_{ki} \\ (N_{kk} - \gamma_k) & N_{ki} \end{bmatrix},$$
(34)

where  $abs(\mathbf{X})$  is the modulus of matrix  $\mathbf{X}$ , *i.e.*, for  $\mathbf{X}=[x_{ij}]$ ,  $abs(\mathbf{X})=[|x_{ij}|]$ .

From the assumptions  $h_n = h_{-n}$ ,  $g_n = -g_{-n}$ ,  $h_{-N/2} = 0$  and from (32), Eqs. (C13) and (C14) are derived in App. C, implying that:

$$2\mathbf{M} = -\mathbf{A}^{-1} \begin{bmatrix} v - u & -v' - u' \\ z + x & z' - x' \end{bmatrix} - \mathbf{B}^{-1} \begin{bmatrix} w + u & w' - u' \\ y - x & -y' - x' \end{bmatrix},$$
(35)

where:

$$\mathbf{M} = \begin{bmatrix} (M_{kk} - 1) & M_{ki} \\ (N_{kk} - \gamma_k) & N_{ki} \end{bmatrix}, \quad \mathbf{A} = \begin{bmatrix} a & b \\ c & d \end{bmatrix}, \quad \mathbf{B} = \begin{bmatrix} e & -b \\ -c & f \end{bmatrix}.$$
(36)

Taking into account (12)-(13), (29)-(31), and (C6)-(C10) from App. C, the properties  $h_n = h_{-n}$ ,  $g_n = -g_{-n}$ ,  $h_{-N/2} = 0$  and  $\alpha_n = \alpha_{-n}$ ,  $\beta_n = -\beta_{-n}$  and considering the range n = -N/2, ..., N/2-1:

$$a = 2\sum_{n} h_n \alpha_n \sin^2 \alpha_n, \quad b = 2\sum_{n} h_n \beta_n \sin \alpha_n \cos \alpha_n, \quad c = 2\sum_{n} g_n \alpha_n \sin \alpha_n \cos \alpha_n, \quad (37)$$

$$d = 2\sum_{n} g_{n} \beta_{n} \cos^{2} \alpha_{n}, \quad e = 2\sum_{n} h_{n} \alpha_{n} \cos^{2} \alpha_{n}, \quad f = 2\sum_{n} g_{n} \beta_{n} \sin^{2} \alpha_{n}, \quad (38)$$

$$u = -2\sum_{n} h_n (\gamma_k \beta_n + \sin 2\gamma_k x_n) \sin \alpha \alpha_n \cos \alpha \alpha_n, \quad v = 2\sum_{n} h_n (\alpha_n - \cos 2\gamma_k x_n) \sin^2 \alpha \alpha_n, \quad (39)$$

$$w = 2\sum_{n} h_n (\alpha_n - \cos 2\gamma_k x_n) \cos^2 \alpha_n, \quad x = 2\sum_{n} g_n (\alpha_n - \cos 2\gamma_k x_n) \sin \alpha_n \cos \alpha_n, \quad (40)$$

$$y = 2\sum_{n} g_n (\gamma_k \beta_n + \sin 2\gamma_k x_n) \sin^2 \alpha_n, \quad z = 2\sum_{n} g_n (\gamma_k \beta_n + \sin 2\gamma_k x_n) \cos^2 \alpha_n.$$
(41)

The relationships for u', v', w', x', y', and z' are determined by (39)-(41) after substituting  $\gamma_i$  for  $\gamma_k$ .

The upper bounds of the component errors of matrix **D** can be obtained by:

$$\mathbf{D}(\tau/R) = \max_{\gamma_k, \gamma_i} \operatorname{abs}\left( \begin{bmatrix} 1 & 0 \\ -\gamma_k/R & 1/R \end{bmatrix} \mathbf{M} \right)$$
(42)

or by:

$$\mathbf{D}(|\lambda_{k} - \lambda_{i}|) = \max_{\substack{\tau, \gamma_{k}, \gamma_{i}:\\|\lambda_{k} - \lambda_{i}| = \text{const.}}} \operatorname{abs}\left(\begin{bmatrix}1 & 0\\ -\gamma_{k}/R & 1/R\end{bmatrix}\mathbf{M}\right) = \max_{\substack{\tau, \gamma_{k}, \gamma_{i}:\\ \tau + \gamma_{i} - \gamma_{k} = \text{const.}}} \operatorname{abs}\left(\begin{bmatrix}1 & 0\\ -\gamma_{k}/R & 1/R\end{bmatrix}\mathbf{M}\right).$$
(43)

The derivation of (42) is presented in [42]; however, the version of the equation presented here differs in the fact that the maximization with respect to  $\gamma_k$  and  $\gamma_i$  is only with respect to the transformation of (34) to (42), *i.e.*, to the last step of the derivation. This also allows for a second method of maximization with respect to  $\gamma_i$  and  $\gamma_k$  for a given  $\tau + \gamma_l - \gamma_k$ , which is performed in (43). This type of maximization indicates the characteristics of the upper bounds of the component errors from matrix **D** as a function of the distance  $|\lambda_k - \lambda_i|$  between the components of the spectrum. These bounds, calculated from (43) for the triangular window  $w_n$ in (9), with  $\eta_1 = 1/2$  and  $\eta_2 = 1/6$  in (6), are presented in Figs. 1 and 2 and are verified using Monte-Carlo simulations in Section 5.



Fig. 1. Upper bounds of the component errors of the total error  $\delta_d |B_k|$  (22) in the estimation of  $|B_k|$  from (33)-(41) for a signal consisting of two complex oscillations, a triangular window  $w_n$  in (9),  $\eta_1=1/2$  and  $\eta_2=1/6$ : a)  $\delta_{dk} |B_k|$ , b)  $\delta'_{di} |B_k|$ .



Fig. 2. Upper bounds of the component errors of the total error  $\Delta_d \lambda_k$  (25) in the estimation of  $\lambda_k$  from (33)-(41) for a signal consisting of two complex oscillations, a triangular window  $w_n$  in (9),  $\eta_1=1/2$ , and  $\eta_2=1/6$ : a)  $\Delta_{dk} \lambda_k$ , b)  $\Delta'_{di} \lambda_k$ .

#### 5. Verification of the LIDFT systematic error equations using Monte Carlo simulations

The upper bounds of the error components from matrix **D** defined by (33) are determined by (43). These bounds are obtained with some simplifying assumptions presented in previous sections, therefore the quality of these approximations must be verified. This verification was performed using simulations in which the samples of a signal consisting of two complex oscillations were processed in the LIDFT algorithm for a given parameter *R* and the value of  $|B_i|/|B_k|$ . Each simulation was performed for  $\lambda_k$ ,  $\lambda_i$ , arg  $B_k$ , and arg  $B_i$  values that were randomly generated from across their full ranges. For this signal, the LIDFT algorithm with a triangular prototype window  $w_n$  with  $\eta_1 = 1/2$  and  $\eta_2 = 1/6$  was applied to obtain the estimates of  $\lambda_k$  and  $|B_k|$ , and the errors of these estimates were calculated. For each combination of the parameters *R* and  $|B_i|/|B_k|$ ,  $10^7$  simulations were performed, and the maximum values of the calculated errors are plotted against the distance  $|\lambda_k - \lambda_i|$  between the components of the spectrum in comparison with the upper bounds determined from (33) and (43), as shown in Fig. 3.



Fig. 3. Comparison of the upper bounds of the total systematic errors from (33) and (43) of the LIDFT method (for the case of the triangular window  $w_n$  in (9),  $\eta_1=1/2$ , and  $\eta_2=1/6$ ) with Monte-Carlo simulation results (with  $10^7$  simulations for each curve) for R = 2, 4, 8 and  $|B_i|/|B_k| = 0.1$ , 10: a) the error of the amplitude estimation,  $\delta_d |B_k|$ ; b) the error of the frequency estimation,  $\Delta_d \lambda_k$ .

The ranges of the plotted errors are limited to the values below 10% for  $\delta_d |B_k|$  (*i.e.*, -20 dB) and below 0.1 bin for  $\Delta_d \lambda_k$  (*i.e.*, -20 dB with respect to 1 bin). Over these ranges, the results presented here converge well to the upper bounds of the errors obtained in Section 4 and the maximum errors of the LIDFT method obtained in the simulations.

# 6. Conclusions

The main results of this paper are analytical formulas for the systematic errors of the LIDFT method and the verification of these formulas using Monte-Carlo simulations. The analysis presented here shows that for multifrequency signals consisting of many complex oscillations, the total systematic errors are the sum of the basic error components ( $\delta_{dk} |B_k|$  for amplitude estimation and  $\Delta_{dk} \lambda_k$  for frequency estimation) and the components of each oscillation ( $\delta_{di} |B_k|$  and  $\Delta_{di} \lambda_k$ , which are proportional to the ratio  $|B_i|/|B_k|$ ), as described by (22)-(27). Supplementing these relations with (32) yields analytical formulas for some particular cases. One of the basic cases is a signal consisting of two complex oscillations, analyzed in Sect. 4. The main results of this section are relations (35)-(41) and (43), which together determine each error component for the case of two complex oscillations (basic components  $\delta_{dk} |B_k|$ ,  $\Delta_{dk} \lambda_k$  and additional components  $\delta'_{di} |B_k|$ ,  $\Delta'_{di} \lambda_k$ , which multiplied by the ratio  $|B_i|/|B_k|$  are the components  $\delta_{di} |B_k|$ ,  $\Delta_{di} \lambda_k$ ). The plots of these four components  $(\delta_{dk} |B_k|, \Delta_{dk} \lambda_k, \delta'_{di} |B_k|, \Delta'_{di} \lambda_k)$  in Figs. 1 and 2 show that increasing the parameter R decreases all of these error components and that the errors increase with small distances between components of the spectrum. The total estimation errors ( $\delta_d | B_k |$  and  $\Delta_d \lambda_k$ ) for the case of two complex oscillations described by (33) are determined based on analytical solutions from Section 4 and verified using Monte-Carlo simulations with randomly generated frequencies and phases of complex oscillations (Fig. 3). Each curve was subjected to  $10^7$ simulations, although far smaller numbers of simulations yielded similar results. These results demonstrate the accuracy of the analytical solutions presented in Section 4 over practical ranges of the errors from Figs. 1-3, *i.e.*, below -20 dB. These results characterize the basic properties of the LIDFT method and its systematic errors, and they can be used to improve this method for the estimation of multifrequency signal parameters.

#### Appendix A. Derivation of (22)-(24)

From (21) and  $M_{kk} \approx 1$  (due to the case of small systematic errors of phase arg  $B_k$  and modulus  $|B_k|$ ):

$$\begin{split} \delta_{d} | B_{k} | &= \max_{\substack{\arg B_{i} \\ i \in S_{1}}} \left| \frac{|\hat{B}_{k}| - |B_{k}|}{|B_{k}|} \right| = \max_{\substack{\arg B_{i} \\ i \in S_{1}}} \left| \frac{|\hat{B}_{k}|}{|B_{k}|} - 1 \right| = \max_{\substack{\arg B_{i} \\ i \in S_{1}}} \left| M_{kk} + \sum_{\substack{i \in S_{1} \\ i \neq k}} M_{ki} \frac{B_{i}}{B_{k}} \right| - 1 \right| \\ &= |M_{kk} - 1| + \sum_{\substack{i \in S_{1} \\ i \neq k}} \frac{|B_{i}|}{|B_{k}|} | M_{ki}| = \delta_{dk} | B_{k} | + \sum_{\substack{i \in S_{1} \\ i \neq k}} \delta_{di} | B_{k} |, \end{split}$$
(A1)

where  $\delta_{dk} |B_k|$  and  $\delta_{di} |B_k|$  are defined by (23) and (24).

# **Appendix B. Derivation of (25)-(27)**

To obtain high accuracy, *i.e.*,  $\delta_d |B_k| \ll 1$ , the condition  $|M_{ki}| \cdot |B_i|/|B_k| \ll 1$  must be fulfilled. Using this condition, (16), (21) and an expansion to a Maclaurin series with respect to  $M_{ki}B_i/B_k$ :

$$\begin{split} \Delta_{d}\lambda_{k} &= \max_{\substack{\operatorname{arg}B_{i}\\i\in S_{1}}}\left|\hat{\lambda}_{k}-\lambda_{k}\right| = \max_{\substack{\operatorname{arg}B_{i}\\i\in S_{1}}}\frac{1}{R}\left|\operatorname{Re}\frac{\hat{C}_{k}}{\hat{B}_{k}}-\gamma_{k}\right| = \max_{\substack{\operatorname{arg}B_{i}\\i\in S_{1}}}\frac{1}{R}\left|\operatorname{Re}\frac{\sum_{i\in S_{1}}N_{ki}B_{i}}{\sum_{i\in S_{1}}M_{ki}B_{i}}-\gamma_{k}\right| \\ &= \max_{\substack{\operatorname{arg}B_{i}\\i\in S_{1}}}\frac{1}{R}\left|\operatorname{Re}\frac{N_{kk}+\sum_{i\in S_{1},i\neq k}\frac{N_{ki}}{M_{ki}}(M_{ki}\frac{B_{i}}{B_{k}})}{M_{kk}+\sum_{i\in S_{1},i\neq k}(M_{ki}\frac{B_{i}}{B_{k}})}-\gamma_{k}\right| \end{split}$$
(B1)  

$$&\cong \max_{\substack{\operatorname{arg}B_{i}\\i\in S_{1}}}\frac{1}{R}\left|\operatorname{Re}\frac{N_{kk}}{M_{kk}}-\gamma_{k}+\operatorname{Re}\sum_{i\in S_{1},i\neq k}\frac{1}{M_{kk}}(N_{ki}-\frac{N_{kk}}{M_{kk}}M_{ki})\right| \\ &= \frac{1}{R}\left|\operatorname{Re}\frac{N_{kk}}{M_{kk}}-\gamma_{k}\right|+\sum_{i\in S_{1},i\neq k}\frac{|B_{i}|}{|B_{k}|}\cdot\frac{1}{R}\left|\frac{1}{M_{kk}}(N_{ki}-\frac{N_{kk}}{M_{kk}}M_{ki})\right|, \end{split}$$

yielding (25) with:

$$\Delta_{dk}\lambda_{k} = R^{-1} \left| \operatorname{Re} \frac{N_{kk}}{M_{kk}} - \gamma_{k} \right| \leq R^{-1} \left| \frac{N_{kk}}{M_{kk}} - \gamma_{k} \right|,$$
(B2)

$$\Delta_{di}\lambda_{k} = \sum_{i\in S_{1}, i\neq k} \frac{|B_{i}|}{|B_{k}|} \cdot \Delta_{di}^{\prime}\lambda_{k} , \qquad \Delta_{di}^{\prime}\lambda_{k} = R^{-1} \left| \frac{1}{M_{kk}} (N_{ki} - \frac{N_{kk}}{M_{kk}} M_{ki}) \right|.$$
(B3)

For small systematic errors  $\delta_{dk} |B_k| \ll 1$ , *i.e.*,  $|M_{kk}-1| \ll 1$ ; thus, the following approximation can be made:

$$M_{kk}^{-1} = [1 + (M_{kk} - 1)]^{-1} = 1 - (M_{kk} - 1) + o((M_{kk} - 1)^2) \approx 1 - (M_{kk} - 1),$$
(B4)

which, when used in (B2)-(B3), using (23)-(24) and under the conditions:

$$\delta_{dk} | B_k | \ll 1, \quad \delta'_{di} | B_k | \ll 1, \quad \Delta_{dk} \lambda_k \ll 1, \quad \Delta'_{di} \lambda_k \ll 1, \tag{B5}$$

yields (26)-(27):

$$\Delta_{dk}\lambda_{k} \cong R^{-1} | N_{kk} - \gamma_{k} - (N_{kk} - \gamma_{k} + \gamma_{k})(M_{kk} - 1) | \cong R^{-1} | (N_{kk} - \gamma_{k}) - \gamma_{k}(M_{kk} - 1) |, \qquad (B6)$$

$$\Delta_{di}^{\prime}\lambda_{k} = R^{-1} \left| \frac{N_{ki}(M_{kk}-1) + N_{ki} - \gamma_{k}M_{ki} - (N_{kk}-\gamma_{k})M_{ki}}{[1 + (M_{kk}-1)]^{2}} \right| \cong R^{-1} \mid N_{ki} - \gamma_{k}M_{ki} \mid.$$
(B7)

# **Appendix C. Derivation of (35)**

For the signal consisting of two complex oscillations  $B_k e^{j2\pi n\lambda_k/N}$  and  $B_i e^{j2\pi n\lambda_i/N}$  defined in the first paragraph of Section 4, (32) has the form:

$$\begin{bmatrix} r_{0} & r_{-\tau} & s_{0} & s_{-\tau} \\ r_{\tau} & r_{0} & s_{\tau} & s_{0} \\ p_{0} & p_{-\tau} & q_{0} & q_{-\tau} \\ p_{\tau} & p_{0} & q_{\tau} & q_{0} \end{bmatrix} \cdot \begin{bmatrix} M_{kk} - \mathbf{I} & M_{ki} \\ M_{ik} & M_{ii} - \mathbf{I} \\ N_{ik} - \gamma_{k} & N_{ki} \\ N_{ik} & N_{ii} - \gamma_{i} \end{bmatrix} = -\begin{bmatrix} \Delta r_{0}(\gamma_{k}) & \Delta r_{-\tau}(\gamma_{i}) & \Delta s_{0}(\gamma_{k}) & \Delta s_{-\tau}(\gamma_{i}) \\ \Delta r_{\tau}(\gamma_{k}) & \Delta r_{0}(\gamma_{i}) & \Delta s_{\tau}(\gamma_{k}) & \Delta s_{0}(\gamma_{i}) \end{bmatrix} \begin{bmatrix} \mathbf{1} & \mathbf{0} \\ \mathbf{0} & \mathbf{1} \\ \gamma_{k} & \mathbf{0} \\ \mathbf{0} & \gamma_{i} \end{bmatrix}.$$
(C1)

The property of an even  $h_n$ , taking into account (6) and (9), means that  $g_n$  is odd. From this assumption and based on (19), function  $H(\lambda)$  is even with respect to  $\lambda$ , and function  $G(\lambda)$  is odd, and taking into account (18):

$$H_{-\tau}(-\gamma) = H_{\tau}(\gamma), \qquad G_{-\tau}(-\gamma) = -G_{\tau}(\gamma). \tag{C2}$$

Condition  $h_{-N/2}=0$  yields the condition  $g_{-N/2}=0$ . From this and from the even and odd properties of  $h_n$  and  $g_n$ , respectively, it follows that the coefficients  $r_m$ ,  $s_m$ ,  $p_m$ , and  $q_m$  from (12)-(13) are real and fulfill the following:

$$r_{-\tau} = r_{\tau}, \ s_{-\tau} = -s_{\tau}, \ p_{-\tau} = -p_{\tau}, \ q_{-\tau} = q_{\tau}, \ s_{0} = p_{0} = 0.$$
(C3)

Analogous conditions are also fulfilled for (29)-(30):

$$\Delta r_{-\tau} = \Delta r_{\tau}, \ \Delta s_{-\tau} = -\Delta s_{\tau}, \ \Delta p_{-\tau} = -\Delta p_{\tau}, \ \Delta q_{-\tau} = \Delta q_{\tau}, \ \Delta s_{0} = \Delta p_{0} = 0.$$
(C4)

Equation (C1), taking into account (C3)-(C4), has the form:

$$\begin{bmatrix} r_{0} & r_{\tau} & 0 & -s_{\tau} \\ r_{\tau} & r_{0} & s_{\tau} & 0 \\ 0 & -p_{\tau} & q_{0} & q_{\tau} \\ p_{\tau} & 0 & q_{\tau} & q_{0} \end{bmatrix} \cdot \begin{bmatrix} M_{kk} - 1 & M_{ki} \\ M_{ik} & M_{ii} - 1 \\ N_{kk} - \gamma_{k} & N_{ki} \\ N_{ik} & N_{ii} - \gamma_{i} \end{bmatrix} = -\begin{bmatrix} \Delta r_{0}(\gamma_{k}) & \Delta r_{\tau}(\gamma_{i}) & 0 & -\Delta s_{\tau}(\gamma_{i}) \\ \Delta r_{\tau}(\gamma_{k}) & \Delta r_{0}(\gamma_{i}) & \Delta s_{\tau}(\gamma_{k}) & 0 \\ 0 & -\Delta p_{\tau}(\gamma_{i}) & \Delta q_{0}(\gamma_{k}) & \Delta q_{\tau}(\gamma_{i}) \\ \Delta p_{\tau}(\gamma_{k}) & 0 & \Delta q_{\tau}(\gamma_{k}) & \Delta q_{0}(\gamma_{i}) \end{bmatrix} \begin{bmatrix} 1 & 0 \\ 0 & 1 \\ \gamma_{k} & 0 \\ 0 & \gamma_{i} \end{bmatrix}.$$
(C5)

To shorten the notations of the next transformations, the 18 symbols (a, b, ...) are introduced as follows:

$$a = r_0 - r_\tau, \ b = -s_\tau, \ c = p_\tau, \ d = q_0 + q_\tau, \ e = r_0 + r_\tau, \ f = q_0 - q_\tau,$$
(C6)

$$u = \gamma_k \Delta s_\tau(\gamma_k), \ v = \Delta r_0(\gamma_k) - \Delta r_\tau(\gamma_k), \ w = \Delta r_0(\gamma_k) + \Delta r_\tau(\gamma_k),$$
(C7)

$$u' = \gamma_i \Delta s_\tau(\gamma_i), \ v' = \Delta r_0(\gamma_i) - \Delta r_\tau(\gamma_i), \ w' = \Delta r_0(\gamma_i) + \Delta r_\tau(\gamma_i),$$
(C8)

$$x = \Delta p_{\tau}(\gamma_k), \ y = \gamma_k (\Delta q_0(\gamma_k) - \Delta q_{\tau}(\gamma_k)), \ z = \gamma_k (\Delta q_0(\gamma_k) + \Delta q_{\tau}(\gamma_k)),$$
(C9)

$$x' = \Delta p_{\tau}(\gamma_i), \ y' = \gamma_i (\Delta q_0(\gamma_i) - \Delta q_{\tau}(\gamma_i)), \ z' = \gamma_i (\Delta q_0(\gamma_i) + \Delta q_{\tau}(\gamma_i)).$$
(C10)

After transformation (C5), by summing the appropriate columns and rows and taking into account (C6)-(C10):

$$\begin{bmatrix} e & a & b & -b \\ e & -a & -b & -b \\ -c & c & d & f \\ c & c & d & -f \end{bmatrix} \cdot \begin{bmatrix} (M_{kk}-1) + M_{ik} & M_{ki} + (M_{ii}-1) \\ (M_{kk}-1) - M_{ik} & M_{ki} - (M_{ii}-1) \\ (N_{kk}-\gamma_{k}) + N_{ik} & N_{ki} + (N_{ii}-\gamma_{i}) \\ (N_{kk}-\gamma_{k}) - N_{ik} & N_{ki} - (N_{ii}-\gamma_{i}) \end{bmatrix} = -\begin{bmatrix} w+v & w'-v'-2u' \\ w-v+2u & w'+v' \\ y+z & z'-y'-2x' \\ z-y+2x & z'+y' \end{bmatrix},$$
(C11)  
$$\begin{bmatrix} e & 0 & 0 & -b \\ 0 & a & b & 0 \\ 0 & c & d & 0 \\ -c & 0 & 0 & f \end{bmatrix} \cdot \begin{bmatrix} (M_{kk}-1) + M_{ik} & M_{ki} - (M_{ii}-1) \\ (M_{kk}-1) - M_{ik} & M_{ki} - (M_{ii}-1) \\ (N_{kk}-\gamma_{k}) + N_{ik} & N_{ki} + (N_{ii}-\gamma_{i}) \\ (N_{kk}-\gamma_{k}) - N_{ik} & N_{ki} - (N_{ii}-\gamma_{i}) \end{bmatrix} = -\begin{bmatrix} w+u & w'-u' \\ v-u & -v'-u' \\ z+x & z'-x' \\ y-x & -y'-x' \end{bmatrix}.$$
(C12)

Omitting elements equal to zero in (C12), two equations are obtained instead of (C5):

$$\begin{bmatrix} a & b \\ c & d \end{bmatrix} \cdot \begin{bmatrix} (M_{kk} - 1) - M_{ik} & M_{ki} - (M_{ii} - 1) \\ (N_{kk} - \gamma_k) + N_{ik} & N_{ki} + (N_{ii} - \gamma_i) \end{bmatrix} = -\begin{bmatrix} v - u & -v' - u' \\ z + x & z' - x' \end{bmatrix},$$
 (C13)

$$\begin{bmatrix} e & -b \\ -c & f \end{bmatrix} \cdot \begin{bmatrix} (M_{kk} - 1) + M_{ik} & M_{ki} + (M_{ii} - 1) \\ (N_{kk} - \gamma_k) - N_{ik} & N_{ki} - (N_{ii} - \gamma_i) \end{bmatrix} = -\begin{bmatrix} w + u & w' - u' \\ y - x & -y' - x' \end{bmatrix}, \quad (C14)$$

and multiplying the left sides of these equations by the inverses of the appropriate matrices and summing by sides yields (35) and (36).

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