A Precise Calculation of Power System Frequency

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Abstract—A precise digital algorithm based on Discrete Fourier Transforms (DFT) to estimate the frequency of a sinusoid with harmonics in real-time is proposed. This algorithm that we called the Smart Discrete Fourier Transforms (SDFT) smartly avoids the errors that arise when frequency deviates from the nominal frequency, and keeps all the advantages of the DFT e.g., immune to harmonics and the recursive computing can be used in SDFT. These make the SDFT more accurate than conventional DFT based techniques. In addition, this method is recursive and very easy to implement, so it is very suitable for use in real-time. We provide the simulation results compared with a conventional DFT method and second-order Prony method to validate the claimed benefits of SDFT.

Index Terms—Discrete Fourier Transforms (DFT), frequency estimation, phasor measurement.

I. INTRODUCTION

F REQUENCY is one of the most important quantities in power system operation because it can reflect the dynamic energy balance between load and generating power. So frequency is always regarded as an index of the operating practices, and utilities can know the system energy balance situations by observing frequency variations. Frequency may vary very fast in the transient events such that it is difficult to track it accurately. In addition, there are many devices, such as power electronic equipments and arc furnaces, etc. generating lots of harmonics and noise in modern power systems. It is therefore essential for utilities to seek and develop a reliable method that can measure frequency in presence of harmonics and noise.

With the advent of the microprocessor, more and more microprocessor-based equipments have been extensively used in power systems. Using such equipments is known to provide accurate, fast responding, economic, and flexible solutions to measurement problems [1]. Therefore, all we have to do is to find the best algorithm and implement it. There have been many digital algorithms applied to estimating frequency during recent years, for example Modified Zero Crossing Technique [2], Level Crossing Technique [3], Least Squares Error Technique [4]–[6], Newton method [7], Kalman Filter [8]–[10], Prony Method [11], and Discrete Fourier Transform (DFT) [12], etc. For real-time use, most of the aforementioned methods have trade-off between accuracy and speed [13]. A precise digital algorithm, namely Smart Discrete Fourier Transform (SDFT) is presented and tries to meet the real-time use. SDFT has the advantages that it can obtain exact solution when frequency

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deviates from nominal frequency, its speed is even faster than DFT, and it can get exact solution in the presence of harmonics.

The organization of this paper is as follows: We describe in very detail the SDFT in Section II. DFT, Prony method and SDFT are tested by three examples in Section III. Finally, we give a conclusion in Section IV.

II. THE PROPOSED DIGITAL ALGORITHM

This section presents the algorithm of the SDFT that calculates the frequency from a voltage/current signal. Consider a sinusoidal input signal of frequency $\omega = 2\pi f$ with *m*th harmonic given by:

$$x(t) = X_1 \cos(\omega t + \phi_1) + X_2 \cos(m\omega t + \phi_2)$$
 (1)

where

 X_1, X_2 : the amplitude,

 ϕ_1, ϕ_2 : the phase angle.

Suppose that x(t) is sampled with a sampling rate (60*N) Hz waveform to produce the sample set $\{x(k)\}$

$$x(k) = X_1 \cos\left(\omega \frac{k}{60N} + \phi_1\right) + X_2 \cos\left(m\omega \frac{k}{60N} + \phi_2\right).$$
(2)

The signal x(t) is conventionally represented by a phasor (a complex number) \overline{x}

$$\overline{x} = Xe^{j\phi} = X\cos\phi + jX\sin\phi.$$
(3)

Then x(t) can be expressed as

$$x(t) = \frac{\overline{x}_1 e^{j\omega t} + \overline{x}_1^* e^{-j\omega t}}{2} + \frac{\overline{x}_2 e^{jm\omega t} + \overline{x}_2^* e^{-jm\omega t}}{2}$$
(4)

where * denotes complex conjugate. Moreover, the fundamental frequency (60 Hz) component of DFT of $\{x(k)\}$ is given by

$$\hat{x}_r = \frac{2}{N} \sum_{k=0}^{N-1} x(k+r) \exp\left(-j\frac{2\pi k}{N}\right).$$
 (5)

Combing (4) and (5) and taking frequency deviation $[\omega = 2\pi(60 + \Delta f)]$ into consideration, we obtain:

$$\hat{x}_r = \frac{\overline{x}_1}{N} \sum_{k=0}^{N-1} \exp\left(j2\pi(60 + \Delta f)\frac{(k+r)}{60N}\right)$$
$$\cdot \exp\left(-j2\pi\frac{k}{N}\right) + \frac{\overline{x}_1^*}{N} \sum_{k=0}^{N-1}$$
$$\cdot \exp\left(-j2\pi(60 + \Delta f)\frac{(k+r)}{60N}\right) \exp\left(-j2\pi\frac{k}{N}\right)$$

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$$+ \frac{\overline{x}_2}{N} \sum_{k=0}^{N-1} \exp\left(j2\pi(60 + \Delta f)\frac{m(k+r)}{60N}\right)$$
$$\cdot \exp\left(-j2\pi\frac{k}{N}\right) + \frac{\overline{x}_2^*}{N} \sum_{k=0}^{N-1}$$
$$\cdot \exp\left(-j2\pi(60 + \Delta f)\frac{m(k+r)}{60N}\right) \exp\left(-j2\pi\frac{k}{N}\right).$$
(6)

We rearrange (6) as the following

$$\hat{x}_{r} = \frac{\overline{x}}{N} \exp\left(j\frac{2\pi}{N}\left(1 + \frac{\Delta f}{60}\right)r\right) \sum_{k=0}^{N-1} \exp\left(j2\pi\frac{\Delta f}{60N}k\right) \quad \text{wf} \\ + \frac{\overline{x}^{*}}{N} \exp\left(-j\frac{2\pi}{N}\left(1 + \frac{\Delta f}{60}\right)r\right) \sum_{k=0}^{N-1} \\ \cdot \exp\left(-j2\pi\frac{\left(2 + \frac{\Delta f}{60}\right)}{N}k\right) \\ + \frac{\overline{x}}{N} \exp\left(j\frac{2\pi}{N}\left(1 + \frac{\Delta f}{60}\right)mr\right) \sum_{k=0}^{N-1} \\ \cdot \exp\left(j2\pi\frac{\left(m - 1 + \frac{m\Delta f}{60}\right)}{N}k\right) \quad \text{Re} \\ + \frac{\overline{x}^{*}}{N} \exp\left(-j\frac{2\pi}{N}\left(1 + \frac{\Delta f}{60}\right)mr\right) \sum_{k=0}^{N-1} \hat{x}_{r} \\ \cdot \exp\left(-j2\pi\frac{\left(-1 - m + \frac{m\Delta f}{60}\right)}{N}k\right). \quad (7)$$

We use the following identity to simplify (7)

$$\sum_{i=0}^{N-1} (e^{j\theta})^i = \frac{\sin\frac{N\theta}{2}}{\sin\frac{\theta}{2}} \exp\left(j(N-1)\frac{\theta}{2}\right). \tag{8}$$

Then (7) can be expressed as

$$\hat{x}_r = \frac{\overline{x}_1}{N} \exp\left(j\frac{2\pi}{N}\left(1 + \frac{\Delta f}{60}\right)r\right) \frac{\sin\frac{N\theta_1}{2}}{\sin\frac{\theta_1}{2}}$$
$$\cdot \exp\left(j(N-1)\frac{\theta_1}{2}\right) + \frac{\overline{x}_1^*}{N} \exp\left(-j\frac{2\pi}{N}\left(1 + \frac{\Delta f}{60}\right)r\right)$$
$$\cdot \frac{\sin\frac{N\theta_2}{2}}{\sin\frac{\theta_2}{2}} \exp\left(j(N-1)\frac{\theta_2}{2}\right) + \frac{\overline{x}_2}{N}$$

$$\cdot \exp\left(j\frac{2\pi}{N}\left(1+\frac{\Delta f}{60}\right)mr\right)\frac{\sin\frac{N\theta_3}{2}}{\sin\frac{\theta_3}{2}}$$
$$\cdot \exp\left(j(N-1)\frac{\theta_3}{2}\right) + \frac{\overline{x}_2^*}{N}$$
$$\cdot \exp\left(-j\frac{2\pi}{N}\left(1+\frac{\Delta f}{60}\right)mr\right)\frac{\sin\frac{N\theta_4}{2}}{\sin\frac{\theta_4}{2}}$$
$$\cdot \exp\left(j(N-1)\frac{\theta_4}{2}\right) \tag{9}$$

where

$$\theta_1 = \frac{2\pi\Delta f}{60N},$$

$$\theta_2 = -\frac{2\pi\left(2 + \frac{\Delta f}{60}\right)}{N},$$

$$\theta_3 = \frac{2\pi\left(m - 1 + \frac{m\Delta f}{60}\right)}{N}, \text{ and }$$

$$\theta_4 = \frac{2\pi\left(-m - 1 + \frac{m\Delta f}{60}\right)}{N}.$$

Rearranging (9) further, we obtain

$$\begin{split} & \cdot = \frac{\overline{x}_1}{N} \frac{\sin \frac{N\theta_1}{2}}{\sin \frac{\theta_1}{2}} \\ & \cdot \exp\left(j\frac{\pi}{60N}(\Delta f(2r+N-1)+120r)\right) \\ & + \frac{\overline{x}_1^*}{N} \frac{\sin \frac{N\theta_2}{2}}{\sin \frac{\theta_2}{2}} \\ & \cdot \exp\left(-j\frac{\pi}{60N}(\Delta f(2r+N-1)+120(r+N-1))\right) \\ & + \frac{\overline{x}_2}{N} \frac{\sin \frac{N\theta_3}{2}}{\sin \frac{\theta_3}{2}} \\ & \cdot \exp\left(j\frac{\pi}{60N}(m\Delta f(2r+N-1)+60(2mr+mN-m-N+1))\right) \\ & + \frac{\overline{x}_2^*}{N} \frac{\sin \frac{N\theta_4}{2}}{\sin \frac{\theta_4}{2}} \\ & \cdot \exp\left(-j\frac{\pi}{60N}(m\Delta f(2r+N-1)+60(2mr+mN-m+N-1))\right). \end{split}$$

(10)

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If we define A_r , B_r , C_r and D_r as

$$A_r = \frac{\overline{x}_1}{N} \frac{\sin \frac{N\theta_1}{2}}{\sin \frac{\theta_1}{2}}$$

$$\cdot \exp\left(j\frac{\pi}{60N}(\Delta f(2r+N-1)+120r)\right) \tag{11}$$

$$B_r = \frac{\overline{x}_1^*}{N} \frac{\sin \frac{N\theta_2}{2}}{\theta_2}$$

$$\sin \frac{\sigma_2}{2} \\ \cdot \exp\left(-j\frac{\pi}{60N}(\Delta f(2r+N-1)+120(r+N-1))\right)$$
(12)

$$C_r = \frac{\overline{x}_2}{N} \frac{\sin \frac{N\theta_3}{2}}{\sin \frac{\theta_3}{2}} \cdot \exp\left(j\frac{\pi}{60N}(m\Delta f(2r+N-1) + 60(2mr+mN-m-N+1))\right)$$
(13)

$$D_{r} = \frac{\overline{x}_{2}^{*}}{N} \frac{\sin \frac{N\theta_{4}}{2}}{\sin \frac{\theta_{4}}{2}} \\ \cdot \exp\left(-j\frac{\pi}{60N}(\Delta f(2mr+N-1) + 60(2mr+mN-m+N-1))\right).$$
(14)

Then (9) can be expressed as

$$\hat{x}_r = A_r + B_r + C_r + D_r.$$
(15)

Except the parts of *m*th harmonic, so far the development of the algorithm of SDFT are the same as the conventional DFT method. So the SDFT can keep all advantages of DFT such as recursive computing manner. But in the DFT, it assumes that the frequency deviation is small enough to be ignored, and it always considers $\hat{x}_r \approx A_r$, so traditional DFT based methods incur error in estimating frequency and phasor when frequency deviates from nominal frequency (60 Hz). If we want to get exact solution, we must take B_r , C_r and D_r into consideration. So we define

$$a = \exp\left(j\left(\frac{\pi}{60N}(2\Delta f + 120)\right)\right).$$
 (16)

And from (10), we will find the following relations

$$A_{r+1} = A_r a \tag{17}$$

$$B_{r+1} = B_r a^{-1} \tag{18}$$

$$C_{r+1} = C_r a^m \tag{19}$$

$$D_{r+1} = D_r a^{-m}.$$
 (20)

$$\hat{x}_{r+1} = A_r a + B_r a^{-1} + C_r a^m + D_r a^{-m} \tag{21}$$

$$\hat{x}_{r+2} = A_{r+1}a + B_{r+1}a^{-1} + C_{r+1}a^m + D_{r+1}a^{-m}.$$
 (22)

If we multiply " a^m " on both sides of (21) and (22), respectively, then we get

$$a^{m}\hat{x}_{r+1} = A_{r}a^{1+m} + B_{r}a^{-1+m} + C_{r}a^{2m} + D_{r}$$
(23)
$$a^{m}\hat{x}_{r+2} = A_{r+1}a^{1+m} + B_{r+1}a^{-1/+m} + C_{r+1}a^{2m} + D_{r+1}.$$
(24)

Subtracting (15) from (23) and subtracting (21) from (24), respectively, we can erase D_r and obtain

$$\hat{y}_{r} = a^{m} \hat{x}_{r+1} - \hat{x}_{r}
= A_{r}(a^{1+m} - 1) + B_{r}(a^{-1+m} - 1) + C_{r}(a^{2m} - 1)$$
(25)

$$\hat{y}_{r+1} = a^{m} \hat{x}_{r+2} - \hat{x}_{r+1}
= A_{r+1}(a^{1+m} - 1) + B_{r+1}(a^{-1+m} - 1) + C_{r+1}(a^{2m} - 1).$$
(26)

Repeat similar operation to erase the C_r and B_r , then the equations will become

$$a\hat{z}_{r+1} - \hat{z}_r = A_r(a^2 - 1)(a^{1-m} - 1)(a^{1+m} - 1)$$
(27)

$$a\hat{z}_{r+2} - \hat{z}_{r+1} = A_{r+1}(a^2 - 1)(a^{1-m} - 1)(a^{1+m} - 1) \quad (28)$$

where $\hat{z}_r = a\hat{u}_{r+1} - \hat{u}_r$, $\hat{u}_r = a^{-m}\hat{y}_{r+1} - \hat{y}_r$. Dividing (28) by (27), we get

$$\frac{a\hat{z}_{r+2} - \hat{z}_{r+1}}{a\hat{z}_{r+1} - \hat{z}_r} = \frac{A_{r+1}}{A_r} = a.$$
 (29)

Then expand (29), and use numerical method to find the solution of "a." And from the definition of "a" in (16), we can get the exact solution of the frequency

$$f = 60 + \Delta f = \cos^{-1}(\operatorname{Re}(a))\frac{60N}{2\pi}.$$
 (30)

From (29), it is observed that SDFT can provide exact frequency using \hat{x}_r , \hat{x}_{r+1} , \hat{x}_{r+2} , \hat{x}_{r+3} and \hat{x}_{r+4} in the presence of harmonics. Moreover, we can estimate phasor after getting exact "f" by the following equations:

$$A_r = \frac{\hat{z}_{r+1}a - \hat{z}_r}{(a^2 - 1)(a^{1-m} - 1)(a^{1+m} - 1)}$$
(31)

$$X_1 = abs(A_r) \frac{N \sin\left(\frac{\pi \Delta f}{60N}\right)}{\sin\left(\frac{\pi \Delta f}{60}\right)}$$
(32)

$$\phi_1 = angle(A_r) - \frac{\pi}{60N} (\Delta f(N-1)).$$
 (33)

It appears that SDFT can take integral order harmonics into consideration. To distinguish easily, SDFT means calculating frequency for m = 1 and we add suffix to the others, for example SDFT₃ and SDFT₃₅ calculate frequency for m = 3 and m = 3, 5, respectively. And here we offer the polynomial equation of SDFT₃ (m = 3):

$$a^4 + p_3 a^3 + p_2 a^2 + p_1 a + p_0 = 0$$

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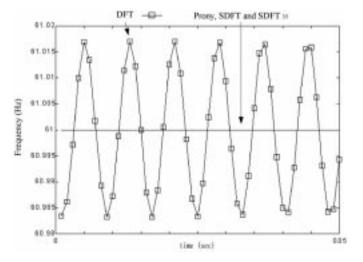


Fig. 1. Comparison of frequency calculations among DFT, Prony, SDFT, and SDFT₃₅. [Test signal: $x(t) = \cos(wt)$, f = 61 Hz].

where

$$p_{3} = -\frac{\hat{x}_{r+1} + \hat{x}_{r+3}}{2\hat{x}_{r+2}}, \qquad p_{2} = -\frac{3}{4},$$

$$p_{1} = \frac{\hat{x}_{r+1} + \hat{x}_{r+3}}{4\hat{x}_{r+2}}, \qquad p_{0} = \frac{\hat{x}_{r} + 2\hat{x}_{r+2} + \hat{x}_{r+4}}{16\hat{x}_{r+2}}.$$

Actually, if we assume that $x(t) = X_1 \cos(\omega_1 t + \phi_1) +$ $X_2 \cos(\omega_2 t + \phi_2)$ from the beginning of development of the algorithm, we will derive a polynomial equation similar to (29) that provide exact frequency in the presence of nonintegral harmonic. We add suffix "n" to SDFT means that has taken nonintegral harmonics into consideration. Although we can take all of the harmonics into consideration, we still need a digital filter to decay noise and high order harmonics. Since, in SDFT, the more harmonics taken into consideration, the more CPU time needed in computing. The advantages of digital filtering are no voltage drop, no temperature drift, no noise addition, and don't have any analog filter element features, like aging. Besides these, digital filter can be implemented in microprocessor-based equipment. These make us choose a digital filter to filter noise and high order harmonics. There are many digital filters that we can choose e.g., Hanning, Hamming and Blackman windows. In our simulations we will use the Blackman window for filtering.

III. SIMULATION RESULTS

Simulation results presented in this section were all simulated from Matlab and showed a fair comparison to both the DFT method and Prony method. In Fig. 1, we showed that SDFT could obtain an exact solution identical to the Prony method under frequency deviation in a pure sinusoidal waveform. Fig. 1 also shows the performance of $SDFT_{35}$ method and conventional DFT method. It is observed that conventional DFT method gives the wrong frequency calculations.

In Fig. 2(a), SDFT₃₅ and SDFT_{3n} are observed to obtain the exact solution. While the SDFT and Prony methods test the same signal without filtering, we find that Prony is worse than SDFT in the presence of harmonics, but if the test signal is filtered by a Blackman window (window size = 16) for estimation,

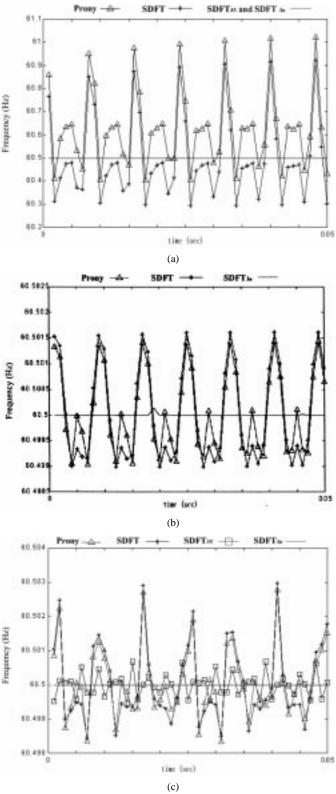


Fig. 2. (a) Comparison of frequency calculations among SDFT, Prony, SDFT₃₅, SDFT_{3n}. [Test signal: $x(t) = \cos(wt) + 0.05\cos(3wt) + 0.02\cos(5wt)$, f = 60.5 Hz]. (b) Comparison of frequency calculations among SDFT, Prony, SDFT_{3n} with Blackman window. (c) Comparison of frequency calculations among SDFT, Prony, SDFT₃₅, SDFT_{3n} with Blackman window. [Test signal: $x(t) = \cos(wt) + 0.05\cos(3wt) + 0.02\cos(2\pi * 380 * t)$, f = 60.5 Hz].

we find that the SDFT and Prony methods have similar performance. Since the $SDFT_{3n}$ can deal with 3rd harmonic and a

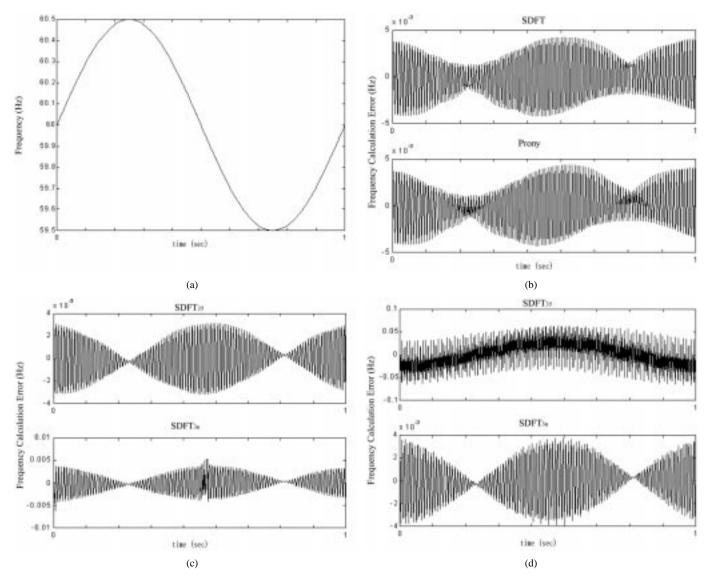


Fig. 3. (a) Frequency variation of test signal, $x(t) = \cos(wt)$, $f = 60 + 0.5 \sin(2\pi t)$ Hz. (b) Comparison of error of frequency calculations between SDFT and Prony. (c) Comparison of error of frequency calculations between SDFT₃₅ and SDFT_{3n}. (d) Comparison of error of frequency calculations between SDFT₃₅ and SDFT_{3n}.

nonintegral harmonic, in Fig. 2(c), only the $SDFT_{3n}$ gets the exact solution.

In Fig. 3(a), the frequency is changed as a sine wave and 3rd, 5th harmonics is also added in test signal during 1 second. We can observe the errors of SDFT and Prony with Blackman window in Fig. 3(b), and the errors of SDFT₃₅ and SDFT_{3n} without filter in Fig. 3(c). Although SDFT₃₅ and SDFT_{3n} can resist the effect of the 3rd, 5th harmonics, the effect of frequency variation makes them get some small errors. In Fig. 3(d), we change 5th harmonic to a nonintegral harmonic, and of course SDFT_{3n} has better performance than SDFT₃₅. However, this is a special case for SDFT_{3n}. In fact, SDFT_{3n} spends more time in computing than SDFT₃₅, and sometimes it has convergence problem when there are more than two harmonics in the signal. Anyway, from Fig. 3 we can conclude that SDFT-family algorithms (SDFT, SDFT₃₅ and SDFT_{3n}) are better than DFT method and Prony method for frequency calculation.

By comparison of computation speed, Table I shows the AMD K6-200 CPU time of each method. There are 960 data

TABLE I COMPUTATION TIME

	Prony	DFT	SDFT	SDFT ₃	SDFT ₃₅	SDFT _{3n}
Time (sec)	2.03	0.71	0.54	1.01	1.71	2.31

per second computed by each method [the test signal is the same as in Fig. 3(a)] without a Blackman window to calculate the frequency, while adding a Blackman window will add 0.91 second to the computation. We find that SDFT is the fastest method in these computations, even faster than DFT, because SDFT counts frequencies directly, but DFT has to count the phase first and then use the phase difference to count frequencies. The faster speed of SDFT over the Prony method is because recursion can be used in SDFT.

IV. CONCLUSION

In this paper we introduce the SDFT-family methods and demonstrate their performance. SDFT both keeps the advantages of DFT and also deals with the cause of frequency deviation errors, while taking harmonics into consideration. These aspects make SDFT a fast, accurate and harmonic-resisting method. But we do not suggest taking all the harmonics into consideration, since that would require too much computation time. Alternatively, using a smoothing window to decay the high order harmonics and just taking the low order harmonics into consideration will be more efficient and suitable for power systems under real-time demands.

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