Performance analysis of low-complexity welch power spectral density for automatic frequency analyser

Teh Yi Jun¹, Asral Bahari Jambek², Uda Hashim³
¹,²School of Microelectronic Engineering, Universiti Malaysia Perlis, Perlis, Malaysia
³Institute of Nano Electronic Engineering, Universiti Malaysia Perlis, Perlis, Malaysia

ABSTRACT

The aim of this paper is to investigate the performance of the Low Complexity Welch Power Spectral Density Computation (PSDC). This algorithm is an improvement from Welch PSDC method to reduce the computational complexity of the method. The effect of the sampling rate and the input frequency toward to accuracy of frequency detection is being evaluated. From the experiment results, sampling rate nearest to the twice of the input frequency provides the highest accuracy which achieved 99%. The ability of the algorithm to perform complex signal also has been investigated.

1. INTRODUCTION

Frequency based sensor have grown rapidly in popularity. Hence, the requirement of frequency analysis also highly increases. In the frequency analysis, the common process is the spectral density calculation. There are many methods of spectral density calculation been proposed, namely as Robust Spectral Density Estimation (RSDE), Independent Component Analysis (ICA), Multitaper Power Spectral Density Estimation (MPSDE), Low-complexity Welch Power Spectral Density (LCWPSD), B-Spline Windows Power Spectral Density Estimation and etc. [1–13]. Among the method listed, LCWPSD is the most suitable for biosensor processing. Hence, it has been proposed to apply in biosensor application. Optimum parameter are required to enhance the performance of LCWPSD. Next sections are going to discuss the details for LCWPSD.

2. LITERATURE REVIEW

In this section, the propose method will be discussed in details on how the spectral density of the signal is detected. LCWPSDC [4] is a modified method from the Welch PSDC [14] in term to reduce the complexity of computational. Figure 1 illustrated the flow of the LCWPSDC. Input signal first will be segmented into them with the length of N. Then, the segment with the length of N will be divided into two windows which length of N/2. After dividing into two windows, the FFT will be applied to each window, the product called N/2-FFT. The N/2-FFT from the window at the same segment will be merge again become N-FFT. The mathematical derivation of FFTs process is present as below.

Journal homepage: http://beei.org/index.php/EEI
The signal of every segment of the input signal as $x[n]$, where the $n=0,…,N-1$ of length of $N$. Given the FFT of $x[n]$ as (1).

$$X(k) = \sum_{n=0}^{N-1} x[n] e^{-j\frac{2\pi nk}{N}} \quad (1)$$

(1) was substituted with $k = s + 2u$ and $n = l + mM$, where $M = N/2$, $s = \{0,1\}$, $u = 0, \ldots$, $M-1$, $m = \{0,1\}$, and $l = 0, 1, \ldots, M-1$, we get ,

$$X_1(k) = \sum_{n=0}^{M-1} x[n] e^{-j\frac{2\pi nk}{M}} \quad (2)$$

$$X_2(k) = \sum_{n=0}^{M-1} x[n + M] e^{-j\frac{2\pi nk}{M}} \quad (3)$$

$$X_3(k) = \sum_{n=0}^{M-1} x[n] e^{-j\frac{\pi n}{M}} e^{-j\frac{2\pi nk}{M}} \quad (4)$$

$$X_4(k) = \sum_{n=0}^{M-1} x[n] e^{-j\frac{\pi n}{M}} e^{-j\frac{2\pi nk}{M}} \quad (5)$$

To merge two N/2-FFT into N-FFT (6) and (7) are applied.

$$X(2u) = X_1(k) + X_2(k) \quad (6)$$

$$X(2u + 1) = X_3(k) - X_4(k) \quad (7)$$

After that the N-FFT will be applied with windowing process. In windowing process, Fractional-delay (FD) finite impulse response (FIR) filter was applied. The coefficients of the FD FIR are determined using (8), where $D$ is the delay with non-integer. In this case, delay is set as 0.5.

$$h[n] = \begin{cases} \text{sinc} \left( n-D - \frac{N}{2} + 1 \right) & 0 \leq n \leq N-1 \\ 0 & \text{otherwise} \end{cases} \quad (8)$$
After windowing process, the periodogram of each segment is determined. The last stage is to calculate the average of the periodogram from all the segments as the PSD of the signal. The LCWPSDC algorithm will simulate thru MATLAB.

3. METHODOLOGY

In Section 2, the LCWPSDC method was discussed. In this section, the experiment will evaluate the effect of the sampling rate and input frequency as well as the ability of algorithm to process complex signal. The experiment will be done using MATLAB simulation. Hence, three experiments will be conducted. The sinusoidal signals are selected as the input signal to perform the experiment. Figure 2 illustrated the original input signal of 20Hz frequency.

![Figure 2. Original input signal of 20Hz frequency](image1)

The first experiment is to determine the relationship between the sampling rate and the accuracy of the algorithm. This experiment used the 20Hz of sinusoidal signal as the input signal. The sampling rate are varied from 50Hz until 250Hz with each increment of 50Hz. Figure 3 illustrated the sampled input signal of 20Hz frequency using 150Hz sampling rate. The second experiment is to determine the relationship between the input frequency and the accuracy of the algorithm. This experiment is set the sampling rate fix at 150Hz while the input sinusoidal signal is varied from 10Hz until 50Hz with each increment 10Hz.

![Figure 3. Sampled input signal of 20Hz frequency using 150Hz sampling rate](image2)

The third experiment is to study the ability of the algorithm to perform frequency detection from complex signal. This experiment is performed with complex input signal which is the summation of 10Hz, 0Hz, and 0.05Hz.
30Hz and 50Hz sinusoidal signal. Figure 4 illustrated the sampled complex input signal of 10Hz, 30Hz and 50Hz using 150Hz sampling rate. Table 1 and Table 2 summarize the experimental setup.

| Table 1. Experimental setup of first experiment | Table 2. Experimental setup of second and third experiment |
|------------------------------------------------|----------------------------------------------------------|
| Input frequency | Sampling Rate | Input frequency | Sampling Rate |
| 20 | 50 |
| 20 | 100 |
| 20 | 150 |
| 20 | 200 |
| 20 | 250 |
| 10 | 150 |
| 20 | 150 |
| 30 | 150 |
| 40 | 150 |
| 50 | 150 |
| Complex signal | 150 |

Figure 4. Sampled complex input signal of 10Hz, 30Hz and 50Hz using 150Hz sampling rate

4. RESULTS AND DISCUSSIONS

As the experiment discussed in Section 3, the results will be discussed in this section. All three experiments results will be discussed in detail. Figure 5 illustrated the spectral density detection that using algorithm LCWPSDC with 150Hz sampling rate for input sinusoidal signal of 30Hz frequency.

Table 3 shows the results of the first experiment to study the effect of sampling rate to the accuracy of the algorithm. From the results, sampling rate at 50Hz is able to achieve 99.17% accuracy of frequency detection whereas sampling rate at 250Hz obtained 96.24% accuracy of frequency detection. The results show that the accuracy are decreases as the sampling rate increases. This trend is observed because sampling rate will use as the scale during FFT and larger scale to measure the value will decrease the accuracy. Hence, the lowest sampling rate provides the highest accuracy and the accuracy is decreasing with the increasing of sampling rate.

Table 4 shows results for the second experiment to study the effect of input frequency to the accuracy of the algorithm. From the results, lowest input frequency with 10Hz has 94.53% accuracy of frequency detection. The highest input frequency with 50Hz has 98.93% accuracy of frequency detection. The results show that in accuracy is increasing while the input frequency is increasing. Base on the Nyquist rate, the lowest sampling rate must be at least twice of input frequency to avoid aliasing [15]. Hence, concluded from the first and second experiment, the sampling rate nearest to twice of input frequency will provide the highest accuracy.

Table 4 shows results for the second experiment to study the effect of input frequency to the accuracy of the algorithm. From the results, lowest input frequency with 10Hz has 94.53% accuracy of frequency detection. The highest input frequency with 50Hz has 98.93% accuracy of frequency detection. The results show that in accuracy is increasing while the input frequency is increasing. Base on the Nyquist rate, the lowest sampling rate must be at least twice of input frequency to avoid aliasing [15]. Hence, concluded from the first and second experiment, the sampling rate nearest to twice of input frequency will provide the highest accuracy.

Figure 6 illustrated the results of the third experiment to study the ability of algorithm on complex signal. From the results, it showed that the algorithm was able to detect the complex signals that consist of summation of 10Hz, 30Hz and 50Hz with the frequencies of 10.55Hz, 30.62Hz and 50.54Hz. This results are compared to Table 4 where when each signal is detected individually for 10Hz, 30Hz and 50Hz input frequencies are 10.55Hz, 30.62Hz and 50.54Hz respectively. This shows that the algorithm detect the complex signal are same as the signal has been detected individually. Hence, it proves that the algorithm are able to perform on complex signal.
Performance analysis of low-complexity welch power spectral density for automatic... (Teh Yi Jun)

5. CONCLUSION

In this paper, the LCPSDC has been reviewed. The effect of the sampling rate of the algorithm to the accuracy of algorithm has been investigated. The relationship of the sampling rate and the accuracy are inversely proportional. The effect of the input frequency to the accuracy of the algorithm also has been investigated. The relationship of the input frequency and the accuracy is directly proportional where the accuracy is higher with the increasing of input frequency. The experiment has determined that LCPSDC has the ability to perform complex signal too. Further experiment will be done to the LCPSDC in future to investigate the optimum parameter for apply in biosensor signal application.

ACKNOWLEDGEMENTS

The authors would like to acknowledge the support from the Fundamental Research Grant Scheme (FRGS) under a grant number of FRGS/1/2014/SG05/UNIMAP/02/3 from the Ministry of Higher Education Malaysia.

REFERENCES

[1] B. Spangl and R. Dutter, “Analyzing short-term measurements of heart rate variability in the frequency domain using robustly estimated spectral density functions,” *Comput. Stat. Data Anal.*, vol. 56, no. 5, pp. 1188-1199, May 2012.
[2] M. Ugur, S. Cekli, and C. P. Uzunoglu, “Amplitude and frequency detection of power system signals with chaotic distortions using independent component analysis,” *Electr. Power Syst. Res.*, vol. 108, pp. 43-49, Mar. 2014.
[3] A. J. Barbour and R. L. Parker, “psd: Adaptive, sine multitaper power spectral density estimation for R,” *Comput. Geosci.*, vol. 63, pp. 1-18, Feb. 2014.
[4] K. K. Parhi and M. Ayinala, “Low-Complexity Welch Power Spectral Density Computation,” *IEEE Trans. Circuits Syst. I Regul. Pap.*, vol. 61, no. 1, pp. 172-182, Jan. 2014.
[5] L. Stanciu and C. Stanciu, “Grouped B-spline windows for power spectral density estimation,” in *International Symposium on Signals, Circuits and Systems ISSCS2013*, 2013, pp. 1-4.
[6] T. Schaffer, B. Hensel, C. W. Ju, and C. Jeleazcov, “Evaluation of techniques for estimating the power spectral density of RR-intervals under paced respiration conditions,” *J. Clin. Monit. Comput.*, vol. 28, no. 5, pp. 481-486, 2014.
[7] N. Balasaraswathy and R. Rajavel, “Low-complexity Power Spectral Density Estimation Power spectral density,” in *Artificial Intelligence and Evolutionary Algorithms in Engineering Systems*, 2015, pp. 273-282.
[8] X.Liu, C.Zhang, Z.Ji, Y.Ma, and X.Shang, “Multiple characteristics analysis of Alzheimer’s electroencephalogram by power spectral density and Lempel – Ziv complexity,” Cogn. Neurodyn., vol. 10, no. 2, pp. 121-133, 2016.
[9] Z.Sun, D.Tian, and X.Ning, “Parameter Estimation Technique for the SNCK Scheme Based on the Spectral-Correlation Density,” Wirel. Pers. Commun., vol. 82, no. 3, pp. 1505-1529, 2015.
[10] R.Wang, J.Wang, H.Yu, X.Wei, C.Yang, and B.Deng, “Power spectral density and coherence analysis of Alzheimer’s EEG,” Cogn. Neurodyn., vol. 9, no. 3, pp. 291-304, 2015.
[11] V.Radha, C.Vimala, and M.Krishnaveni, “Power Spectral Density Estimation Using Yule Walker AR Method for Tamil Speech Signal,” Inf. Syst. Indian Lang., pp. 284-288, 2011.
[12] A.Nasser, A.Mansour, K. C.Yao, H.Abdallah, and H.Charara, “Spectrum sensing based on cumulative power spectral density,” EURASIP J. Adv. Signal Process., 2017.
[13] G.Huang, J.Meng, D.Zhang, and X.Zhu, “Window Function for EEG Power Density Estimation and Its Application in SSVEP Based BCIs,” in International Conference on Intelligent Robotics and Applications, 2011, pp. 135-144.
[14] P.Welch, “The use of fast Fourier transform for the estimation of power spectra: A method based on time averaging over short, modified periodograms,” IEEE Trans. Audio Electroacoust., vol. 15, no. 2, pp. 70-73, Jun. 1967.
[15] J.W.Leis, Digital Signal Processing Using MATLAB for Students and Researchers. Hoboken, NJ, USA: John Wiley & Sons, Inc., 2011.

BIOGRAPHIES OF AUTHORS

Teh Yi Jun is currently a PhD student in School of Microelectronics Engineering, Universiti Malaysia Perlis. His main research is on nano-bio sensors signal characteristics and its on-chip analysis algorithms for early diseases detection. He has his B. Eng. degree in Microelectronic engineering from Universiti Malaysia Perlis in 2014.

Dr. Asral Bahari is a senior lecturer at the School of Microelectronics Engineering, Universiti Malaysia Perlis (UniMAP), and was a Programme Chairperson for the Electronics Engineering Degree Programme, UniMAP, from September 2009 to Mac 2013. He has more than 15 years’ experience in VLSI design in both the industry and academic sectors, and has been involved at various levels of VLSI design such as transistor modelling, digital circuit design, analogue circuit design, logic synthesis and physical place and route, architecture design and algorithm development.

Prof. Dr. Uda Hashim is a Professor at Institute of Nano Electronic Engineering (INEE), Universiti Malaysia Perlis (UniMAP), and was a Director for the INEE, UniMAP since October 2008. He has more than 25 years experience in Semiconductor Devices, CMOS Based Sensor, Nanoelectronic and Nano Biochip in both the industry and academic sectors.