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SPECTRAL AVERAGING TECHNIQUES FOR UNSYNCHRONIZED WAVEFORMS انقضيات لتقدير المتوسط الطيفى لموجات فيرا متزامنــــة

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الخلامة - بـ بـ يـعرض هذا - الـبـحث طربـقــُـين لـحـاب الـمـُـوسط -الـطيفى لـمجموعة -مــــــــــــــ الموجات الغلز متزامنة : طريقة اقراد طيف الطور وطربقة متجه الطور ، ويقللون البحث بين أداً * الطريقتين من حيث تحلين نسبة الأشارة " = " الى " = الصوفـــــــــاً * والحصول على شكل تمط متوسّط لمجموعة السّوجات الداخلة في حداب المتوسطاً - اوقــــد .
أظهرت النتائج أن الطريقة الثانية أحمن في الأداء وأنعب للحصول على تقديـــــــ غير مشره لشكل الموجة الاطبية ا.

ABSTRACT-Two algorithms for averaging unsynchronized waveforms are considered; the phase unwrapping and the phase vector techniques. The two techniques employ Fourier analysis and are applied to simulated signals contaminated with random noise. They both achieve a reasonable signal-to-noise ratio
improvement and yield an average waveform that is properly
positioned close to the ensemble mean delay. However, the phase vector method is shown to have a better performance and to be more convenient for producing an undistorted estimate of the original waveform.

I. INTRODUCTION

Signal averaging is often employed to recover a vaveform in the presence of noise. In many cases, it is desired to
obtain a representative pattern for an ensemble of vaveforms
embedded in noise or to determine the frequency content
associated with a particular vaveform once the have been eliminated. When this averaging is taken in the time domain on an ensemble of vaveforms which are synchronized with respect to a time reference, the result will be a faithful
signal average. In this process, the coherence between signal
components will tend to enhance the signal during accumulation components will tend to emance the signal during accomutation
while the random noise components will, tend to cancel one
another. Generally, the signal-to-noise ratio will improve at a
rate proportional to the square root

the lack of a synchronizing pulse. Alternative
cross-correlation techniques; template matching and template
updating techniques are usually employed with unsynchronized
data. These often depend on an initial guess of the w is a low signal-to-noise ratio and especially if the signal and

 $E - 13$

noise have similar spectral densities.

A more desirable technique for averaging waveforms in the absence of an external synchronizing signal would be one which averages the frequency spectra for the ensemble, for both phase as well as magnitude. Motivated by the relationship between
phase and delay, Cheung [3] have developed a spectral averaging technique which yields an estimate of the signal at its mean delay. The technique involves Fourier transformation follved by an averaging scheme whereby all harmonics contained within the amplitude and phase spectra are averaged. Transformation back
to the time domain is then performed. This technique has been
videly used in situations where it is difficult to achieve
synchronization for an ensemble of vave signals $[4,5]$.

This paper presents two algorithms for performing spectral averaging namely; the phase unwrapping and the phase vector methods. The capability of improving the signal-to-noise ratio for each algorithm is determined through examples of simulated signals containing noise levels comparable to those encountered in blological signals. The efficiency of the two algorithms to
yield an estimate of the signal which is positioned close to the ensemble mean delay is also compared using an ensemble of simulated compound action potential (CAP) vaveforms.

II. ENSEMBLE AVERAGE AND THE JITTER PROBLEN

If a deterministic signal $s(t)$ is corrupted by additive
noise $n(t)$ and there is an ensemble of vaveforms $\{r_i(t)\}\,$,

 $r_{1}(t) = s(t) + n_{1}(t), \qquad i=0,...,M,$ $0 \le t \le T$ (1)

then the average of the ensemble of vaveforms can be used as an estimate of the signal

> $y(t) = \frac{1}{M} \sum_{i=1}^{M} r_i(t),$ $0 \leqslant t \leqslant T$ (2)

Applications of this technique in biological signal analysis are numerous and include the estimation of an average locomotor electromyographic pattern (6), electrogastographic vaveforms [7] and electrocardiographic (ECG) beats [8].

It can be shown [1] that if n(t) is a zero-mean stationary process, uncorrelated from waveform to another and uncorrelated with s(t), then the ensemble average forms a consistent signal estimator, i.e.

$$
E\left\{y(t)\right\} = s(t)
$$

Var
$$
\left\{y(t)\right\} = \frac{1}{H} \text{ var } \left\{n(t)\right\}
$$
 (3)

where $E(y(t))$ is the expected value and var($y(t)$) is the variance of y at instant t. Hence, it may be concluded that

the signal-to-noise ratio improves with a factor $M^{1/2}$. Ensemble averaging can also be evaluated from a filtering
standpoint. Rompelman and Ross (1) have reported that the equivalent filter transfer function is

$$
H(f) = \frac{1}{M} \frac{\sin M\pi fT}{\sin n fT}
$$
 (4)

where T is the waveform interval.

However, the estimation of the signal by ensemble averaging is affected by variations in signal delay; ensemble average in the presence of signal jitter results in a
distortion that can also be interpreted from a filtering viewpolnt.

Consider a signal waveform $s(t)$ which is subject to a random jitter δ_i . For simplicity, let us assume δ_i to be uniformly distributed between $(-6, 6)$. By ensemble averaging, the signal is

$$
\tilde{s}(t) = \frac{1}{N} \sum_{i=1}^{M} s(t + \delta_i)
$$
 (5)

As M tends to infinity, this is equivalent to convolving s(t) with the distribution function P(6). This is the same as multiplying the spectrum of s(t) with that of P(6) and hence causes smoothing. It is obvious that the wider the distribution function P(6) (that is, the larger time jitter), the sharper the filtering effect becomes.

If in addition to the jitter, there is a significant
amount of noise, a difficult allgnment problem arises. Some researchers have usel cross-correlation techniques [2,9] where at each stage, an updated signal estimate is used as a template to realign the ensemble. The convergence of these techniques
depends on the nature of the signal, signal-to-noise, and on the initial signal estimate. An alternative approach adopted by some researchers utilizes the FFT transformation to determine a good estimate of the signal at its mean delay in the ensemble of waveforms. This is known as spectral averaging.

III. SPECTRAL AVERAGING

A simple time delay manifests itself as an additive linear phase in the frequency domain. Using this relationship between phase and time delay, a frequency domain averaging has been
developed. Given an ensemble of noisy waveforms in which the signal has a variable delay, spectral averaging yields an estimate of the signal at its mean delay. Two different methods $E-16$

are folloved to implement this averaging: the phase-unwrapping method and the phase vector method.

III.1 Phase-Unwrapping Method

ō

Consider an ensemble of (noise-free) waveforms

 $\tau_i(t) = s(t - D_i), \quad i = 0, ..., M, \quad 0 \leq t \leq T$ (6) where D_i , is the delay. The Fourier transform of $r_i(t)$ is

$$
R_{i}(\omega) = |R_{i}(\omega)| \exp(j\phi_{r1}(\omega)), \qquad i=1,...,H \qquad (7a)
$$

$$
\mathbf{where} \qquad \left| R_{1}(\omega) \right| = \left| s(\omega) \right| \tag{7b}
$$

and
$$
\phi_{ri}(\omega) = \phi_{ci}(\omega) - \omega D_i
$$
 (7c)

and where ϕ represents the Fourier phase. If we could now use these phases , $\phi_{r,i}(\omega)$, to compute ϕ_{sD}^- where

$$
\phi_{\text{S}} \overline{D} \left(\omega \right) = \phi_{\text{S}} \left(\omega \right) - \omega \overline{D} \tag{8a}
$$

and

$$
= \frac{1}{M} \sum_{i=1}^{N} D_i
$$
 (8b)

we could then reconstruct the signal at its mean delay (10),

$$
s(t - \vec{D}) = F^{-1}\left\{ s(\omega) - exp(j\phi_{\vec{SD}}(\omega)) \right\}
$$
 (9)

where F^{-1} stands for the inverse Fourier transform of the quantity in brackets.

Hovever, we can not simply average the principal phase values of the measurements $\phi_{r,i}(\omega)$ to produce $\phi_{sD}(\omega)$ since, in general, averaging principal values gives a biased estimate of the principal value of the average; although the phase
components calculated from the discrete Fourier transform can have values in the range $[-\infty, \infty]$, the values of $\phi_{r_i}(t)$ are all mapped in the region $[-\pi,\pi]$. This is due to the periodic
character of the arctg function. Therefore, we must use the unwrapped phase, a function which is reliably obtained for the discrete Fourier transform (DFT) by a procedure due to Cheung (3). The procedure consists of adding or subtracting multiples
of 2π to the phase values at appropriate frequencies in order
to make all the differences of two consecutive phase spectral
values less than π . The remo spectrum reveals any trend which may be present. This trend represents the displacement of the main signal features from their temporally most-symmetrical position. Therefore, the unwrapping procedure must be followed by a linear regression

and a trend removal. The whole process is known as "Regression Spinning" [11].

Thus, with $\phi_{r\, \mathrm{i}}$ now representing the unwrapped detrended Fourier phase, we can write

$$
\phi_{\text{S}D}(\omega) = \frac{1}{M} \sum_{i=1}^{M} \phi_{\text{r}i}(\omega) \tag{10}
$$

which represents the average phase value that can be used in reconstructing the spectral average using Eq. (9).

III.2 The Phase Vector Method

An alternative phase averaging procedure has been adopted by Sayers et al. (12). Consider a discrete signal x(n) and its
discrete Fourier transform X(k). X(k) is a complex quantity having real and imaginary parts R(k) and I(k)

$$
X(k) = R(k) + I(k)
$$
 (11)

The amplitude $A(k)$ and phase $\phi(k)$ of each spectral component are defined as:

$$
A(k) = \sqrt{R^2(k) + I^2(k)}
$$
 (12)

$$
\phi(k) = \arctg\left\{\frac{1(k)}{R(k)}\right\} \tag{13}
$$

The average Fourier amplitude is computed as

$$
\overline{\mathbf{A}}(k) = \frac{1}{M} \sum_{i=1}^{M} \mathbf{A}_{i}(k)
$$
 (14)

The average phase is formed by the normalized vector method. A normalised phase vector is defined as:

$$
X(k) = \left\{ \frac{R(k) + I(k)}{A(k)} \right\} = C(k) + j S(k)
$$
 (15)

The average C(k) and S(k) of real and imaginary parts of the M-various phase vectors are:

$$
\overline{C}(k) = \frac{1}{M} \sum_{i=1}^{M} C_i(k), \qquad \overline{S}(k) = \frac{1}{M} \sum_{i=1}^{M} S_i(k) \qquad (16)
$$

 $E-17$

Then the mean phase is obtalned as the arctangent of the ratio $\overline{S}(k)$ to $\overline{C}(k)$

$$
\vec{\phi}(k) = \text{arctg} \left\{ \frac{S(k)}{\vec{C}(k)} \right\}
$$
 (17)

At the end of the averaging process, the average amplitude 15 combined with the average phase angle. Each harmonic is
considered independently and the result of an inverse discrete Fourier transform would yield the spectral average.

IV. SIMULATIONS

Two types of simulated signals have been used to compare between the above averaging algorithms. The first type consists of two hundred identical single cycle discrete sine signals of length 128 samples. Each sequence is given a random delay
(phase shift) derived from uniform distribution of 64 units
vide and centered at time unit 64. Random numbers uniformly
distributed between 0 and 1 have been genera generator. A Gaussian distributed sequence has then been
generated according to an algorithm suggested in (13). The composite signals $r_i(t)$ given in Eq. (1) are formed by adding the Gaussian white noise to signal s(t) such that the signal-to-noise ratio (SNR) is

> SNR = A / σ (18)

where A is the peak of the sine wave and σ is the standard deviation of the noise. Fig.1 shows examples of the signal $r_i(t)$ with SNR of 5. Fig.2 shows a typical record with its

corresponding amplitude and phase spectra.

The second class of simulated waveforms is an ensemble of 100 waveforms simulating the waveforms of the compound action potential (CAP) of the auditory nerve response to click stimulus. This signal is used in electrocochleography (ECochG) for otological and audiological examination. It represents responses from individual primary auditory neurons due to a stimulus.

modeled by a linear segmental The CAP waveforms are approximation; linear segments linking the extremum points in
each vaveform. The duration and slope of the linear segmental
representation are evaluated. The signal has been simulated
using the information derived from the distributions of the original records [14]. It is possible to generate series of random numbers derived from a gaussian distribution using the mean value and twice the value of the corresponding standard deviation of the durations and slopes. For each specific slope, a duration of random value is assigned until an interval of 64 samples (10 msec) has been accumulated

to produce one pseudo-record bearing the major features of the original CAP patterns. The resultant signal is then smoothed using a Hanning filter in order bot obtain the mearest shape (Fig. 3). It is clear that the original waveforms have different latencies, i.e. unsynchronized.

V. RESULTS

Fig.4 shows the ensemble average waveform for the 200 simulated synchronized vaveforms (SNR = 5 and ϕ_i = 0). As can be

seen, in coherent averaging all signals are symmetrically added, whereas the random noise components are summed and tend to be reduced as the number of vaveforms is increased. The figure also illustrates the amplitude and unwrapped phase spectra of the synchronized ensemble average. It is clear that the ngise has been eliminated to a large extent.

The results of ensemble averaging of the unsynchronized
vaveforms with uniformly distributed delay are shown in Fig.5.a. It is clear that the signal jitter is severe enough to obscure much of the signal details (i.e. waveform distortion). The amplitude of the frequency harmonics is decreased but the phase is still unchanged. Fig.5.b shows the ensemble averaged CAP's. It is clear that the pattern main features has been altered due to the variable latencies of the original vaveforms.

Fig.6.(a,b) shows the spectral average pattern using the phase unvrapping method. Magnitude and phase estimates vere computed as in (11) and (12) and signal recovery performed as in (13). The procedure allows waveforms to be positioned close to certaln mean delay but does not improve the SNR as much as the coherent average (Fig. 4).

The results of the phase vector method are shown in Fig. 7. (a,b). It is clear that SNR is higher than that obtained from the phase unwrapping method. Moreover, the CAP waveform is positioned closer to the mean delay of the ensemble.

The above results have been validated and confirmed by
calculating the correlation coefficient (CC) between the average obtained from each algorithm and each waveform of the corresponding ensemble. As the original vaveforms have
different latencies, the sum of the calculated CC's are
expected to be zero if the resultant average Is exactly positioned at the mean delay of the ensemble of vaveforms. The CC's is found to have an average value of 0.0088 for the phase unwrapping method and 0.00094 for the unit phasor method in case of the sine waveforms (Fig.1). The corresponding values of the CAP waveforms are: - 0.342 for the phase unvrapping method and 0.00518 for the unit phasor technique. These values of the correletion cofficients confirm that the unit phasor method gives more accurate average pattern than that obtained from the phase unwrapping technique.

VI. CONCLUSION

Two algorithms for signal averaging of unsynchronized data have been investigated in the present work; the phase

unwrapping and the phase vector methods. The two techniques have improved the signal-to noise ratio of repetitave signals buried in noise. They both employ the Fourier transform and phase averaging to achieve synchronization. However, the phase vector algorithm achieves better SNR improvement and yields an average pattern that is positioned closer to the ensemble mean latency. The results obtained are important for the evaluation of an average pattern for an ensemble of noisy vaveforms when it is difficult to achieve synchronization.

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Flg.1 Examples of simulated noisy signals (SNR =5)

Fig. 2.a- A typical simulated waveform
b- its amplitude spectrum c- the corresponding unwrapped phase spectrum

Fig. 3 Examples of CAP waveforms

Fig. 4 The resultant average of 200 synchronized signals a- ensemble average

- b- amplitude spectrum
- c- unwrapped phase spectrum
- d- SNR versus the number of averaged patterns

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Fig.6 Spectral average using the phase unwrapping method: a- noisy sinusoidal signals
b- simulated CAP signals

Fig.6 Cont.

Fig.7 Spectral average pattern using phase vector technique:
a- noisy slnusoidal signals
b- simulated CAP signals

 -13.5