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► **To cite this version:**

Gabriel Rilling, Patrick Flandrin, Paulo Gonçalves, Jonathan M. Lilly. Bivariate Empirical Mode Decomposition. 10 pages, 3 figures. Submitted to Signal Processing Letters, IEEE. Matlab/C codes and additional .. 2007. <ensl-00137611>

HAL Id: ensl-00137611

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Submitted on 20 Mar 2007

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Bivariate Empirical Mode Decomposition

Gabriel Rilling, Patrick Flandrin, *Fellow, IEEE*, Paulo Gonçalves, Jonathan M. Lilly

Abstract

The Empirical Mode Decomposition (EMD) has been introduced quite recently to adaptively decompose nonstationary and/or nonlinear time series [1]. The method being initially limited to real-valued time series, we propose here an extension to bivariate (or complex-valued) time series which generalizes the rationale underlying the EMD to the bivariate framework. Where the EMD extracts zero-mean oscillating components, the proposed bivariate extension is designed to extract zero-mean rotating components. The method is illustrated on a real-world signal and properties of the output components are discussed. Free Matlab/C codes are available at <http://perso.ens-lyon.fr/patrick.flandrin>.

Index Terms

Empirical Mode Decomposition, complex-valued signals, bivariate time series

EDICS Category: DSP-TFSR

I. INTRODUCTION

In its original formulation [1], the Empirical Mode Decomposition (EMD) can only be applied to real-valued time series. The purpose of this paper is to introduce a new extension of the EMD destined to handle bivariate (or complex-valued) time series. Note however that not all bivariate time series can be processed by this new method but only those where the two components can be assimilated to Cartesian

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coordinates of a point moving in a 2-dimensional space. In particular, the meaning of the signal should not depend on the choice of such Cartesian coordinates. It is worth noticing that another bivariate extension has been introduced very recently [2]. The difference with the one we propose here is significant, but in a nutshell that other method cleverly uses the original EMD to decompose bivariate time series, while ours is a new algorithm that adapts the rationale underlying the EMD to the bivariate framework. Further comparison of the approaches however is out of the scope of this paper. The communication is organized as follows. The bivariate extension is introduced in Section II. Section III is about the components of the resulting decomposition and an illustration is proposed in Section IV. Additionally, free Matlab/C codes corresponding to the proposed algorithms are made available at <http://perso.ens-lyon.fr/patrick.flandrin> along with small scripts aimed at reproducing the figures and other EMD-related software.

II. FROM UNIVARIATE EMD TO BIVARIATE EMD

A. Classical EMD

Basically, the EMD considers a signal at the scale of its local oscillations. The main idea of EMD is then to formalize the idea that, locally: “signal = fast oscillations superimposed on slow oscillations”. Looking at a single oscillation (defined, e.g., as the signal between two consecutive local minima), the EMD is designed to define a local “low frequency” component as the *local trend* $m_1[x](t)$, supporting a local “high frequency” component as a zero-mean oscillation or *local detail* $d_1[x](t)$, so that we can express $x(t)$ as

$$x(t) = m_1[x](t) + d_1[x](t). \quad (1)$$

By construction, $d_1[x](t)$ is an oscillatory signal and, if it is furthermore required to be locally zero-mean everywhere, it corresponds to what is referred to as an *Intrinsic Mode Function* (IMF) [1]. Practically, this primarily implies that all its maxima are positive and all its minima are negative. On the other hand, all we know about $m_1[x](t)$ is that it locally oscillates more slowly than $d_1[x](t)$. We can then apply the same decomposition to it, leading to $m_1[x](t) = m_2[x](t) + d_2[x](t)$ and, recursively applying this on the $m_k[x](t)$, we get a representation of $x(t)$ of the form

$$x(t) = m_K[x](t) + \sum_{k=1}^K d_k[x](t). \quad (2)$$

The discrimination between “fast” and “slow” oscillations is obtained through an algorithm referred to as the *sifting process* [1] which iterates a nonlinear elementary operator \mathcal{S} on the signal until some

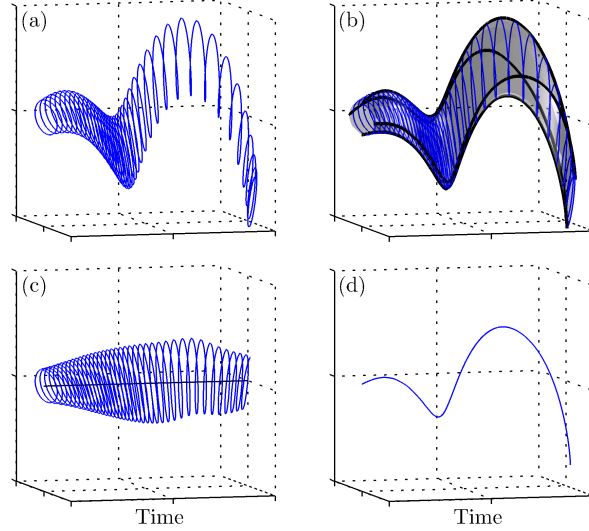


Fig. 1. The principle of the bivariate extensions. (a) A composite rotating signal. (b) The signal enclosed in its 3D envelope. The black thick lines stand for the envelope curves that are used to derive the mean. (c) Rapidly rotating component. (d) More slowly rotating component corresponding to the mean of the tube in (b).

stopping criterion is met. Given a signal $x(t)$, the operator \mathcal{S} is defined by the following procedure:

- 1 Identify all extrema of $x(t)$
- 2 Interpolate (using a cubic spline) between minima (resp. maxima), ending up with some “envelope” $e_{min}(t)$ (resp. $e_{max}(t)$)
- 3 Compute the mean $m(t) = \frac{(e_{min}(t)+e_{max}(t))}{2}$
- 4 Subtract [to] from the signal to obtain $\mathcal{S}[x](t) = x(t) - m(t)$

If the convergence criterion is met after n iterations, the *local detail* and the *local trend* are defined as $d_1[x](t) = \mathcal{S}^n[x](t)$ and $m_1[x](t) = x(t) - d_1[x](t)$.

B. Envelopes in 3 dimensions

The EMD is based on the intuitive notion of “oscillation” which naturally relates to local extrema. But the notion of oscillation is much more confusing when the analyzed data is intrinsically bivariate and it is unclear how to define and interpret local extrema. What is rather clear on the other hand is the notion of rotation, which moreover is arguably a two-dimensional extension of the usual notion of a univariate oscillation. Therefore, the basic idea underlying the proposed bivariate EMD is to formalize the following idea: “bivariate signal = fast rotations superimposed on slower rotations”. As with the classical EMD,

it is clear that the adopted viewpoint is a priori rather restrictive as, e.g., a white noise signal is not meaningfully treated as a sum of oscillations (or rotations). Still this does not prevent the algorithm from producing a decomposition for any signal, as with the univariate EMD.

In order to separate the more rapidly rotating component from slower ones, the idea is once again to define the slowly rotating component as the mean of some “envelope”. Yet the envelope is now a 3-dimensional tube that tightly encloses the signal (see Fig. 1 (b)). Given this, the slowly rotating portion of the signal at any point in time can then be defined as the center of the enclosing tube. To this end, only a given number of points on the tube’s periphery are considered, each one being associated with a specific direction. If only 4 points are used, these can be the extreme points in the directions top, bottom, left and right (see Fig. 2). In practice, the top point, for example, is uniquely defined only when the signal reaches a local maximum in the vertical direction and is therefore tangent to the top of the tube. Between such characteristic moments in time, the top point is then simply defined using interpolation, ending up with the “envelope” associated with the upwards direction (cf the black thick lines in Fig. 1 (b)). Now, given some set of points on the tube periphery at a given instant in time, there are at least two ways to define their mean:

- 1) define the mean as the barycenter of the 4 points, considering each to have unit mass (see Fig. 2 (a)).
- 2) define the mean as the intersection of two straight lines, one being halfway between the two horizontal tangents, the other one halfway between the vertical ones (see Fig. 2 (b)).

In practice, however, the second scheme may be preferred because it is naturally more robust to sampling errors. More precisely, the reason for this is that the envelope points are defined up to an uncertainty that is not isotropic. Indeed, the order of magnitude of this uncertainty can be estimated through a Taylor expansion, which results in an uncertainty jointly proportional to dx/dt and to the sampling period. Thus, the uncertainty is greatest in the direction locally tangent to the signal, and much smaller (of second order) in the orthogonal direction. As the second scheme only uses information from the orthogonal direction, it is naturally more accurate, especially when the signal is sampled sparsely with respect to its period. Note that sampling effects shall not be taken too lightly as the original EMD has been shown to be very sensitive to sampling[3].

The desired goal concerning the interpolation is the same as for the classical EMD: a smooth interpolation with as few “spurious bumps” as possible. Among common interpolation schemes, this calls for cubic spline as it is well known for its minimum curvature property and, in practice, it is still considered the best interpolation scheme for the EMD[4].

In the preceding discussion, we have limited ourselves to 4 directions for the sake of simplicity,

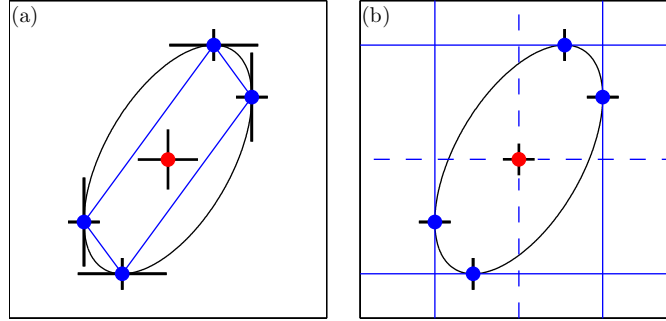


Fig. 2. Definition of the mean of the envelope for each algorithm. The accuracies of the estimation of the sampling points and of their mean are represented schematically as the thick lines of variable length under each point. (the provided script allows to easily test other configurations)

but there is of course no such limitation in practice. Moreover a large number of directions may be interesting insofar as it reduces the dependance of the final decomposition with respect to rotations of the spatial coordinates. For convenience of the presentation, the bivariate time series in the following will be treated as complex-valued time series. Given a set of directions $\varphi_k = 2k\pi/N$, $1 \leq k \leq N$, the proposed bivariate extensions are defined by the same algorithm as the basic EMD, only with new sifting elementary operators \mathcal{S}^{B1} and \mathcal{S}^{B2} corresponding respectively to the algorithms [Algo. 1](#) and [Algo. 2](#).

Algorithm 1: EMD bivariate extension: scheme 1

- 1 **for** $1 \leq k \leq N$ **do**
 - 2 Project the complex-valued signal $x(t)$ on direction φ_k :

$$p_{\varphi_k}(t) = \text{Re} (e^{-i\varphi_k} x(t))$$
 - 3 Extract the locations $\{t_i^k\}$ of the maxima of $p_{\varphi_k}(t)$.
 - 4 Interpolate the set $\{(t_i^k, x(t_i^k))\}$ to obtain the
envelope curve in direction φ_k : $e_{\varphi_k}(t)$.
 - 5 Compute the mean of all envelope curves: $m(t) = \frac{1}{N} \sum_k e_{\varphi_k}(t)$
 - 6 Subtract the mean to obtain $\mathcal{S}^{B1}[x](t) = x(t) - m(t)$
-

Furthermore, the second algorithm can be greatly simplified if we notice that the set that is interpolated at step 4 is in fact included in the plane containing the time axis and direction φ_k . Thus, the interpolation that is performed is very similar to that in the original EMD sifting. Hence, if the number of considered

Algorithm 2: EMD bivariate extension: scheme 2

- 1 **for** $1 \leq k \leq N$ **do**
 - 2 Project the complex-valued signal $x(t)$ on direction φ_k :

$$p_{\varphi_k}(t) = \text{Re}(e^{-i\varphi_k} x(t))$$
 - 3 Extract the maxima of $p_{\varphi_k}(t)$: $\{t_i^k, p_i^k\}$.
 - 4 Interpolate the set $\{(t_i^k, e^{i\varphi_k} p_i^k)\}$ to obtain the partial envelope curve in direction φ_k : $e'_{\varphi_k}(t)$.
 - 5 Compute the mean of all tangents: $m(t) = \frac{2}{N} \sum_k e'_{\varphi_k}(t)$
 - 6 Subtract the mean to obtain $\mathcal{S}^{B2}[x](t) = x(t) - m(t)$
-

directions is an even number, the second algorithm can be expressed in terms of the sifting operator for the univariate EMD (see [Algo. 3](#)). From a theoretical viewpoint, this is interesting as it allows to study the behavior of the algorithm in the light of what is already known about the classical EMD.

Algorithm 3: Reformulation of scheme 2

- 1 **for** $1 \leq k \leq N/2$ **do**
 - 2 Project the complex-valued signal $x(t)$ on direction φ_k :

$$p_{\varphi_k}(t) = \text{Re}(e^{-i\varphi_k} x(t))$$
 - 3 Compute the partial estimate in direction φ_k :

$$s_{\varphi_k}(t) = e^{i\varphi_k} \mathcal{S}[p_{\varphi_k}](t)$$
 - 4 Compute the final estimate: $\mathcal{S}^{B2}[x](t) = \frac{2}{N} \sum_k s_{\varphi_k}(t)$
-

III. BIVARIATE INTRINSIC MODE FUNCTIONS

The proposed EMD bivariate extensions have been designed so that signals rotating around zero are admissible outputs. As this is a rather vague notion, the purpose of this section is to clarify what signals the algorithms actually consider admissible outputs, that is, we ask what sort of signal $x(t)$ is nearly a fixed point of the sifting operator:

$$\mathcal{S}^{B1}x(t) \approx x(t) \quad \text{or} \quad \mathcal{S}^{B2}x(t) \approx x(t). \quad (3)$$

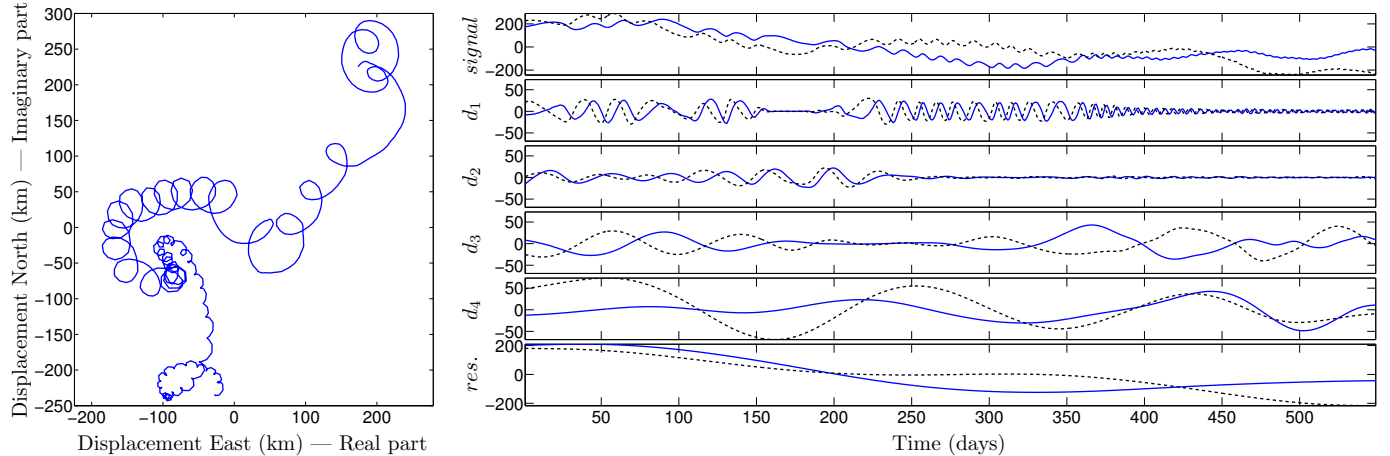


Fig. 3. A signal and its Bivariate Empirical Mode Decomposition. The real parts are plotted as solid blue lines and the imaginary parts as dashed black lines. The decomposition has been obtained with the second algorithm using 64 directions and 10 iterations to extract each component. Portions where components are rotating can be identified by a constant phase shift between the real and imaginary parts: if these are in quadrature, then the signal is rotating circularly; other phase shifts correspond to elliptic rotations.

For the sake of simplicity, we will only address the case of simple periodic solutions which are convenient to describe while still being rather general. Indeed, the method operating at a local scale, its behavior on signals whose properties evolve slowly with respect to the local period is very similar to its operation on exactly periodic signals of constant amplitude. Moreover, signals will here also be considered infinitely continuously differentiable, since the purpose of this section is to provide insight into some possible solutions rather than to make an exhaustive study.

What we find from various simulations is that both algorithms generally accept two types of solutions, the first one corresponding to rotating signals as intended and the second one to cases where the method actually fails to extract rotating components and therefore outputs signals that wander around zero in a more complicated way. It is worth noting that examples of the second type of solutions are generally encountered when the analyzed signal does not clearly contain rotating components, as in e.g. a complex-valued white gaussian noise signal. Moreover, it seems that these non-rotating solutions almost never occur when the analyzed signal is analytic (or anti-analytic). On the other hand, solutions from the first type are signals in which the local sense of rotation never changes. The latter can be defined, e.g., from the sense of the vector product of the velocity and acceleration vectors:

$$\text{sign} \left\{ \text{Im} \left\{ \frac{dx}{dt} \cdot \left(\frac{d^2x}{dt^2} \right)^* \right\} \right\}. \quad (4)$$

Alternatively, if the signal rotates in a counterclockwise fashion (and $|dx/dt| > 0, \forall t$), this exactly means that the signal's derivative can be expressed as

$$\frac{dx}{dt}(t) = r(t)e^{i\psi(t)}, \quad r(t) > 0, \quad \frac{d\psi(t)}{dt} > 0. \quad (5)$$

Finally, periodic signals satisfying (5) are not necessarily fixed points of the sifting operators (3). One of the reasons for this is that there is no condition specifying that they are rotating *around zero* yet. In order to clarify what is meant by “around zero”, we can consider the very simple case in which the signal performs only one full rotation around zero per period ($\psi(t+T) = \psi(t) + 2\pi$ where T is the period). This property simplifies greatly the study, as one can easily show that it implies there is only one maximum per period at step 2 in the algorithms. Therefore, all envelope curves are in fact constants with respect to time, which allows one to derive the mean analytically. The envelope curve associated with the direction $\varphi_k = 2k\pi/N$ is then equal to the maximum signal value in that direction, where the phase of the signal's derivative is $\psi(t) = 2k\pi/N + \pi/2$. Thus, the mean for the first algorithm reads

$$m^{B1}(t) = \frac{1}{N} \sum_k x \left(\psi^{-1} \left(\frac{2k\pi}{N} + \frac{\pi}{2} \right) \right). \quad (6)$$

In the limit where the number of directions tends to infinity, this results in

$$m^{B1}(t) = \frac{1}{2\pi} \int_0^{2\pi} x(\psi^{-1}(\varphi)) d\varphi = \frac{1}{2\pi} \int_0^T x(t) \frac{d\psi}{dt} dt, \quad (7)$$

which is simply the mean of the signal over a period weighted by $d\psi(t)/dt > 0$, where the weighting conveys the fact that the distribution of sampling points on the tube section is denser where the curvature is larger. Likewise, the same reasoning for the second algorithm results in

$$\begin{aligned} m^{B2}(t) &= \frac{1}{\pi} \int_0^T e^{i\psi(t)} \operatorname{Re} \left(e^{-i\psi(t)} x(t) \right) \frac{d\psi}{dt} dt, \\ &= m^{B1}(t) + \frac{1}{2\pi} \int_0^T e^{2i\psi(t)} x^*(t) \frac{d\psi}{dt} dt, \end{aligned} \quad (8)$$

and hence $m^{B2}(t) = m^{B1}(t)$ since

$$\begin{aligned} &\int_0^T e^{2i\psi(t)} x^*(t) \frac{d\psi}{dt} dt \\ &= \frac{i}{2} \left(\left[-e^{2i\psi(t)} x^*(t) \right]_0^T + \int_0^T \frac{dx^*}{dt} e^{2i\psi(t)} dt \right) \\ &= \frac{i}{2} \int_0^T r(t) e^{i\psi(t)} dt = \frac{i}{2} \int_0^T \frac{dx}{dt} dt = 0 \end{aligned} \quad (9)$$

Thus, in this very simple case, the mean is in fact the same for both algorithms and therefore such a simple signal is a fixed point of both sifting operators iff the integral (7) is close to zero.

More generally, the outputs of the two proposed algorithms are very similar when the data clearly contains rotating components, but they may differ significantly when they fail to extract rotating components. Notice, though, that rarely one method succeeds in retrieving rotating components when the other one fails.

IV. ILLUSTRATION

A typical application of the proposed algorithms is proposed in Fig. 3. The data is a position record from an acoustically tracked, neutrally buoyant subsurface oceanographic float, one of a number deployed in the eastern subtropical North Atlantic Ocean in order to track the motion of dense salty water flowing out from the Mediterranean Sea during the “Eastern Basin” experiment [5]. The data is available online from the World Ocean Circulation Experiment Subsurface Float Data Assembly Center (WFDAC) at <http://wfdac.who.i.edu>. Looping trajectories are indicative of intense swirling currents around an isolated packet of Mediterranean Sea water. Such structures, called “coherent vortices”, are frequently observed in the ocean [6] and are more generally a ubiquitous feature of rotating turbulent fluids [7]. Applied to such signals which a priori contain meaningful rotating components, the output of the bivariate extensions typically provide the given decomposition, where the rotations that were apparent in the original signal have been isolated in separate components. As mentioned earlier, though, not all components are rotating but primarily the first two ones, which correspond to the presumed coherent vortex. On this example however, we do not precisely know what information can be extracted from the decomposition yet, but we expect the large scale non-rotating components to reveal useful information regarding the background fluctuations determining the vortex position while the rotating components can a priori be used to extract finer informations, such as amplitude, angular frequency, etc... Such advanced study of the rotating components has already been performed using wavelet ridges to extract the coherent vortex signal [8]. A comparative study of the EMD bivariate extensions for similar purposes is currently under investigation.

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