The Fast Resampled Iterative Filtering Method

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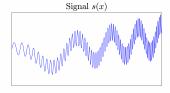
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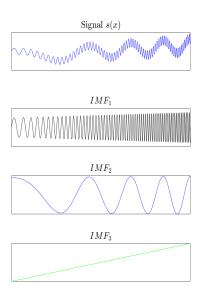
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Iterative Filtering

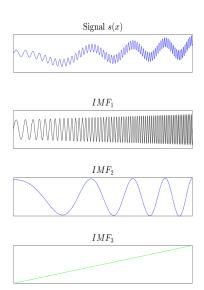
Empirical Method Decomposition (EMD)



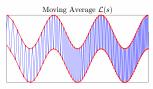
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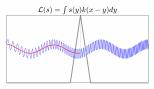
Empirical Method Decomposition (EMD)



The effect of the moving average is to flatten the highest frequency component



A way to emulate the effect is to use a filter on the signal



Iterative Filtering

Choose the filter k:

- Unit-norm, even, nonnegative and compact supported
- $k = \omega \star \omega$ $\implies 0 < \hat{k}(\xi) < 1$

 $s = s - S^{\infty}(s)$

The IF method iteratively apply the filter through convolution
$$\mathcal{S}(f) := f(x) - \int f(y) k(x-y) dy$$

$$IMF = IMF \cup \{\mathcal{S}^{\infty}(s)\}$$

On the Time Dimension the Sifting Operator is the difference between the signal and the Moving Average and it extracts the higher frequencies.

This and the convergence of $S^{\infty}(s)$ can be studied on the frequencies space

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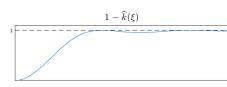
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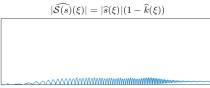
On the Frequency Domain

$$\widehat{\mathcal{S}(s)}(\xi) = \widehat{s}(\xi)(1 - \widehat{k}(\xi))$$
 $\widehat{\mathcal{S}^m(s)}(\xi) = \widehat{s}(\xi)(1 - \widehat{k}(\xi))^m$



 $|\widehat{s}(\xi)|$





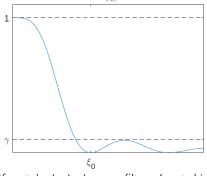
The Fundamental Zero and the Stopping Criterion

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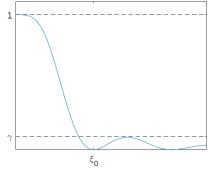


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If we take $k \star k \star k \star k \star ...$ as filter, J_0 gets bigger and the filter gets smoother

Theorem (B. 2023)

If we choose ξ_0 depending on the biggest frequency in \hat{s} whose intensity is at least η , then

$$B(\xi_0, C \sqrt[2p]{\eta \delta}) \subseteq J_0$$

where 2p is the order of ξ_0 the first zero in \hat{k} , and δ depends on the stopping criterion

Bigger J_0 achieves better decomposition, especially for amplitude-modulated signals

$$s(x) = a(x)g(x) \implies \widehat{s}(\xi) = (\widehat{a} \star \widehat{g})(\xi)$$

where a(x) has low instant frequency

Discrete Setting

The signal s(x) is studied on [0,1] and it is supposed to be periodic at the boundaries [Stallone, Cicone, Materassi 2020] so that the discretization results in a circulant matrix

$$s = [s(h) \ s(2h) \dots s(1-h) \ s(1)] \qquad h = 1/N$$

$$S(s)(x) = s(x) - \int_0^1 s(x-y)k(y)dy|_{x=ah} \qquad S(s)(ah) \sim s_a - \frac{1}{N} \sum_{b=1}^N k(bh) s_{a-b}$$

$$S(s) := s - Ks = (I - K)s$$

$$f(s) := s - Ks = (I - K)s$$

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One can thus write the main loop of the discrete IF Algorithm as S(f) := (I - K)f

$$\begin{split} \textit{IMF} &= \textit{IMF} \cup \{\mathcal{S}^m(s)\} \\ s &= s - \mathcal{S}^m(s) \\ \text{where the stopping condition is } \|\mathcal{S}^{m+1}(s) - \mathcal{S}^m(s)\| < \delta \end{split}$$

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$$IMF = IMF \cup \{S^{m}(s)\}$$

$$s = s - S^{m}(s)$$

where the stopping condition is $\|\mathcal{S}^{m+1}(s) - \mathcal{S}^m(s)\| < \delta$

Fast IF

$$\mathcal{S}^m(s) = (I - \mathcal{K})^m s \implies \widehat{\mathcal{S}^m(s)} = k^{\circ m} \circ \widehat{s}$$

where k is the first row of l - K, \circ is the elementwise product and \hat{s} is the DFT of s

$$\|\mathcal{S}^{m+1}(s) - \mathcal{S}^m(s)\| < \delta \iff \|k^{\circ m} \circ (k-e) \circ \widehat{s}\| < \delta$$

The stopping condition can be checked on k and \hat{s} with linear cost + 2 DFT per IMF

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Theorem

If k is a filter, then $0 \le k \le 1$, so $\mathcal{S}^m(s)$ always converges

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Theorem (Cicone, Zhou, 2021, B. 2023)

Given $\delta > 0$, s, then

$$\frac{m^m}{(m+1)^{m+1}}<\frac{\delta}{\|s\|}$$

implies
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Theorem (B. 2023)

 $||IMF_i|| > \eta$

$$\widehat{\mathit{IMF}}_j = \lambda_j \circ \widehat{s}$$

where $0 \le \lambda_j$ and $\sum_j \lambda_j \le 1$. Thus, there is a finite number of relevant IMF, i.e.

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For any vectors h, s let K be any $n \times n$ Hermitian matrix with spectrum in [0,1]. Then

$$\|S^m(s+h)-S^m(s)\|\leq \|h\|.$$

If now the filters and m_j are fixed, for IMF_j the modes generated by s and for IMF_j^* generated by s+h, we have $\sum \|IMF_j^* - IMF_j\|^2 \le \|h\|^2.$

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 and $\sum \lambda_i \leq 1$. Thus, t

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If now the filters and m_i are fixed, for **IMF**; the modes generated by **s** and for

IMF^{*}_i generated by s + h, we have

 $\|\mathcal{S}^m(s+h)-\mathcal{S}^m(s)\|<\|h\|.$

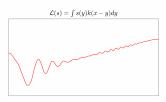
 $\sum \|\mathbf{IMF}_{j}^{*} - \mathbf{IMF}_{j}\|^{2} \leq \|\mathbf{h}\|^{2}.$

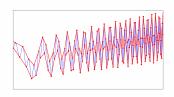
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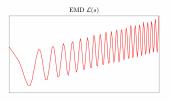
The approximation error of IMF; with respect to the continuous algorithm modes IMF; is proportional to $\log(1/\delta)/n$

Drawbacks





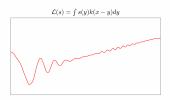


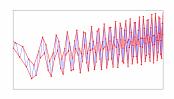


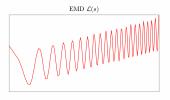
What's happening? Let's take a look at the instantaneous frequencies

Drawbacks



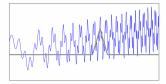


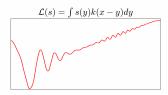


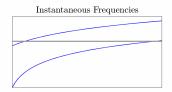


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Drawbacks







$$\widehat{\mathcal{L}(s)}(\xi) = \widehat{s}(\xi) \cdot \widehat{k}(\xi)$$

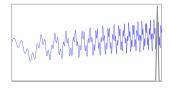
IF does not work with non-disjoint bands of frequencies

Adaptive Local Iterative Filtering

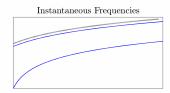
Adaptive Local Iterative Filtering

$$k_x(y) := k(\ell(x)^{-1}y)\ell(x)^{-1}$$

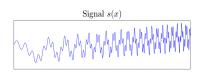
$$k_x(y) := k(\ell(x)^{-1}y)\ell(x)^{-1}$$
 $S(s)(x) := s(x) - \int s(y)k_x(x-y)dy$



$$\mathcal{L}(s) = \int s(y)k_x(x-y)dy$$

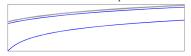


Adaptive Local Iterative Filtering



$$\mathcal{L}(s) = \int s(y) k_x(x-y) dy$$





Given the signal s(x), fix the filter

$$k_{x}(y) := k(\ell(x)^{-1}y)\ell(x)^{-1}$$

where ideally $\ell(x) \sim \xi_0/f(x)$, with f(x) being the instantaneous frequency of the higher-frequency IMF.

Apply iteratively the filter through sifting $S(f) := f(x) - \int f(y)k_x(x-y)dy$ $IMF = IMF \cup \{S^{\infty}(s)\}$ $s = s - S^{\infty}(s)$

ALIF is now as flexible as EMD, and empirically converges, but..

- No structure, not fast as IF (O(n²) against
 O(n log(n)))
- Has no clean formal analysis since it is not a convolution
- $S^{\infty}(s)$ is not always convergent (in the discrete setting) even with a stopping condition

Discrete ALIF and SALIF

$$\mathcal{S}_{ALIF}(s)(x) = s(x) - \int_0^1 s(y) k_x(x-y) dy|_{x=ah} \quad \sim \quad \boldsymbol{s}_a - \frac{1}{N} \sum_{b=1}^N k \left(\frac{(a-b)h}{\ell(ah)} \right) \frac{1}{\ell(ah)} \boldsymbol{s}_b$$

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$$S_{ALIF}(s) := s - Ks = (I - K)s$$

- $\mathcal{S}^{\infty}_{ALIF}(s)$ converges when $|\lambda_i(I-K)| < 1 \, ee \, \lambda_i(I-K) = 1$
- Converges to the kernel of K
- K may have sparse [B., Cicone 2022] negative eigenvalues, so the convergence is not always assured

Discrete ALIF and SALIF

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$$|\lambda_i(I-K)| < 1 \lor \lambda_i(I-K) = 1$$

- Converges to the kernel of K
- K may have sparse [B., Cicone 2022] negative eigenvalues, so the convergence is not always assured
- $S_{SALIF}(s) := s K^T K s = (I K^T K) s$
 - K^TK Has the same kernel of K
 - converges • The method is way slower: the cost

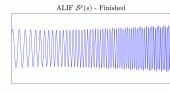
• $1 > \lambda_i(K^T K) \ge 0$ so $S_{SALIF}^{\infty}(s)$ always

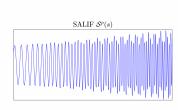
per iteration is doubled and the eigenvalues are closer to zero, so it's

harder to extract the components

 $T \times 1/20$

N = 3000





$$S(s) = (I - K^T K)s$$
 $1 \ge \lambda_i(K^T K) \ge 0$

Since $||K^TK|| \le 1$ and it is Hermitian, we can recover some of the IF good properties:

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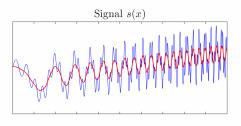
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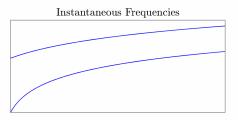
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Resampling





Resampling Function G(y)

Recall that in ALIF the length $\ell(x)$ is computed as $\xi_0/f(x)$ where f(x) is the highest instantaneous frequency for the IMFs of the signal s(x). From now on $\xi_0 = 1$.

Example: The Instantaneous Frequency of $s(x) = \cos(\alpha(x))$ is $\alpha'(x)$ if it is regular enough. In this case, $\ell(x) = 1/\alpha'(x)$.

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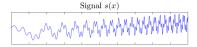
In the Resampled IF (RIF), we instead operate a IF loop to the resampled stationary signal s(G(y)) where

$$G^{-1}(z) = \int_0^z \frac{1}{\ell(x)} dx$$

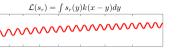
Example: In the previous example, $G^{-1}(z) = \int_0^z \alpha'(x) = \alpha(z) - \alpha(0)$ so that $s(G(y)) = \cos(\alpha(G(y))) = \cos(\alpha(0) + y)$

is a stationary signal with frequency equal to $\xi_0=1$

Resampled Iterative Filtering



Resampled Signal
$$s_r(x)$$



Resampled Moving Average

Given the signal s(x), compute the resampling

$$s_r(x) := s(G(x))$$
 $G^{-1}(z) = \int_0^z \frac{1}{\ell(x)} dx$ and apply iteratively the filter through convolution $S(f) := f(x) - \int f(y) k(x-y) dy$

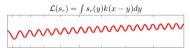
$$S(t) := f(x) - \int f(y)k(x - y)dy$$

$$IMF = IMF \cup \{S^{\infty}(s_r)(G^{-1}(x))\}$$

$$s = s - S^{\infty}(s_r)(G^{-1}(x))$$

Resampled Iterative Filtering

Resampled Signal
$$s_r(x)$$



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 $G^{-1}(z)=\int_0^z rac{1}{\ell(x)}dx$ and apply iteratively the filter through convolution

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$$IMF = IMF \cup \{S^{\infty}(s_r)(G^{-1}(x))\}$$

$$s = s - S^{\infty}(s_r)(G^{-1}(x))$$

We have an algorithm that is

- As flexible as ALIF and SALIF
- Efficient as Fast IF, the resampling is outside the iterations and has the same complexity as the FFT, thus way faster than ALIF and SALIF
- Differently from ALIF, $S^{\infty}(s_r)$ is always convergent because it is an IF iteration. In particular, given a stopping criterion with $\delta > 0$ we have the same results that limit the number of iterations

Theorem

Given $0 < \hat{k} \le 1$, $\delta > 0$, $s_r(x) \in L^2(\mathbb{R})$, then

$$\frac{m^m}{(m+1)^{m+1}}<\frac{\delta}{\|s_r\|}$$
 implies $\|\mathcal{S}^{m+1}(s_r)-\mathcal{S}^m(s_r)\|<\delta$

Theorem

For any $h, s_r \in L^2$

 $\|\mathcal{S}^m(s_r+h)-\mathcal{S}^m(s_r)\|<\|h\|$

 $\widehat{\mathcal{S}}^m(\mathbf{s}_r) = \mathbf{k}^{\circ m} \circ \widehat{\mathbf{s}}_r$ $\|\mathcal{S}^{m+1}(s_r) - \mathcal{S}^m(s)_r\| < \delta \iff \|\mathbf{k}^{\circ m} \circ (\mathbf{k} - \mathbf{e}) \circ \widehat{s_r}\| < \delta$ The stopping condition is checked on k and $\hat{s_r}$ with linear cost + 2 DFT



Theorem

Given $0 < \hat{k} < 1$, $\delta > 0$, $s_r(x) \in L^2(\mathbb{R})$, then

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Theorem

For any $h, s_r \in L^2$ $\|S^{m}(s_{r}+h)-S^{m}(s_{r})\|<\|h\|$

- We don't know if we can still recover
 - Global perturbation results
 - Intrinsic relation with \hat{s}
 - Limited number of meaningful IMFs

Non-Stationary Error Bounds

Let us suppose that the signal s(x) is a linear combination of non-stationary components

$$s(x) := \sum_{j=1}^{M} a_j g_j(x) \qquad g_j(x) = \cos(\alpha_j(x)), \qquad |a_j| \le P$$

$$s_r(z) := \sum_{j=1}^{M} a_j h_j(z) \qquad h_j(z) = \cos(\alpha_j(\alpha_1^{-1}(2\pi sz))) = \cos(\beta_j(z)) \qquad h_1(z) = \cos(2\pi sz)$$

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Theorem (B. 2023)

Let
$$\beta : \mathbb{R} \to \mathbb{R}$$
 be a C^1 function with $\beta'(x) \in [a,b]$ 1-periodic, $0 < a < b$, $R := b-a$. Let $f(x) := \cos(\beta(x))$ and let $f(x)_N$ be the N-tail of its Fourier series, and $G := 2\pi N - b > 0$

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$$\|f(x)_N\|_2^2 \le \min\left\{\left(\frac{b}{G + b + 2\pi}\right)^2, \frac{R^2}{\pi^3 G}\right\}$$

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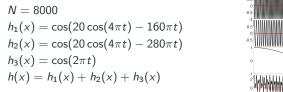
If now
$$j > 1$$
, $f(z) = h_j(z)$ and $N = s - 1$, then $P||f(x) - f(x)_N||_2$ is a bound on the perturbation of the IMF caused by the j -th component h_i , and it is proportional to both

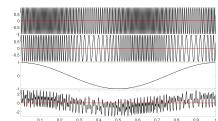
 $\frac{b}{G+b+2\pi} = \frac{\max_{z} \beta_{j}'(z)}{2\pi s} = \max_{x} \frac{\alpha_{j}'(x)}{\alpha_{1}'(x)} \quad \text{(low for far frequencies)}$

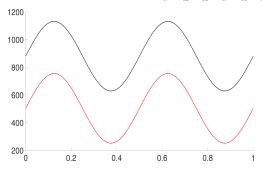
 $R = \max_{z} \beta_j'(z) - \min_{z} \beta_j'(z) = 2\pi s \left(\max_{x} \frac{\alpha_j'(x)}{\alpha_1'(x)} - \min_{x} \frac{\alpha_j'(x)}{\alpha_1'(x)} \right) \quad \text{(zero if same shape)}$

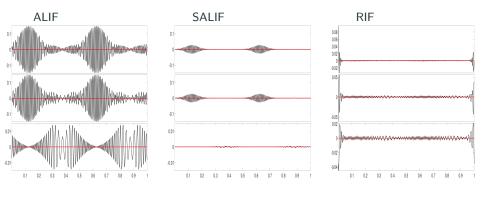
Numerical Experiments

Experiment 1



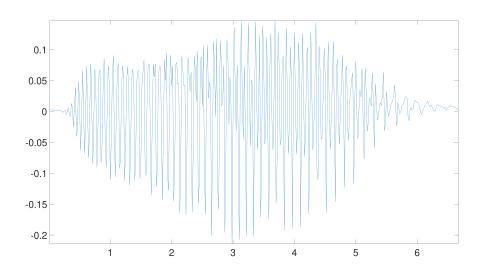




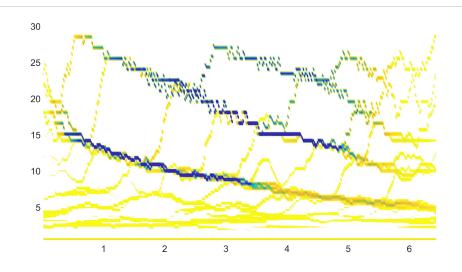


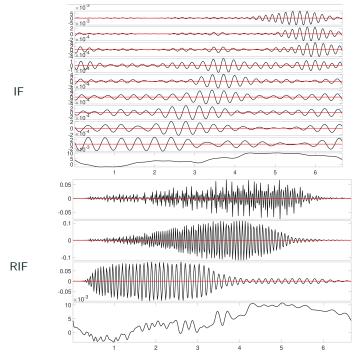
	Time	err1	err2	err3	Niter1	Niter2
ALIF	4.0860	0.070388	0.071158	0.008549	18	2
SALIF	19.7919	0.010054	0.010055	0.000161	353	5
RIF	1.4724	0.070388 0.010054 0.003426	0.003292	0.000908	81	11

Experiment 2



Experiment 2





Conclusions and Future Works

We developed Algorithms and Theory for

- SALIF Stable, Flexible, Convergent but very Slow
- RIF Flexible, Convergent, Fast but may introduce inaccuracies

Moreover RIF proves himself also Aliasing-Free and we also expanded the theory of IF.

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We developed Algorithms and Theory for

- SALIF Stable, Flexible, Convergent but very Slow
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Moreover RIF proves himself also Aliasing-Free and we also expanded the theory of IF.

Still to do:

- Better exploit the order of zero of the filter
- Further analysis of IF for non-stationary and AM components
- We can use RIF to better study ALIF through the relation between G(x) and $\ell(x)$
- Better ways to compute G(x) without relying on $\ell(x)$
- · Improve the error bounds, since they prove to be empirically better
- How perturbation affect the output of RIF

Thank You!





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