

# COHERENT NOISE REMOVAL IN SEISMIC DATA WITH REDUNDANT MULTISCALE DIRECTIONAL FILTERS

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## ABSTRACT

Directional filters are a commonly used tool in modern seismic data processing to process coherent signals depending on their apparent slowness or slope. This operation enhances the characterization of the great variety of signals present in a seismic dataset that enables a better characterization of the subsurface structure. This paper compares two complementary local adaptive multiscale directional filters: a directional filter bank based on dual-tree  $M$ -band wavelets and a novel local slant stack transform (LSST) based filter in the time-scale domain. Their differences resides in redundancy levels and slope (directional) resolution. To provide an objective measurement of their quality, the structural similarity index measure has been employed to compare the results obtained with both approaches on a real seismic dataset example.

## 1. INTRODUCTION AND MOTIVATIONS

The complexity of seismic data processing has contributed to the development of several efficient signal processing tools such as wavelet transforms or spike deconvolution. While 1-D processing of time sampled signals (*e.g.* seismic traces) is relatively common in geophysics, the assemblage of signals into 2-D images on a spatial axis (*e.g.* offset) related to sensor location, as represented in Figure 1, opens access to more involved two-dimensional processing tools. Yet, differences between traditional and seismic images foster the quest for specific adaptations or genuine developments. Seismic images could be described as band-pass images in the time direction (between 5 and 80 Hz typically), with high lateral (along offsets) semblance, fairly comparable to fingerprints with reduced isotropy. We refer to the following freely available book [6] for a combination of signal analysis and seismic processing topics.

The application under consideration is a seismic equivalent to occlusion removal. Coherent noises are caused by peculiar wave propagation. They arise as structured signals in seismic records, hampering subsequent geophysical data processing and interpretation with high amplitude directional band-pass stripes. This work investigates their removal with two breeds of selective multiscale directional decompositions. They further allow to enlighten some similarities between traditional and geophysical data processing that deserve further investigation, for the benefit of signal processing.

The two features most commonly used to detect and characterize the great variety of signals that seismic record sections convey are:

- the similarity along signal trajectory,

- the velocity/slowness vector of the seismic waves at the receiver.

When the seismic trace (1-D signal) density is high, the high similarity enables the design of a great variety of filters depending on the signal slope, to increase the signal-to-noise and the signal-to-interference ratios.

The modern signal processing tools used in this field root in the classical plane-wave decomposition techniques. We can classify these classical techniques in several broad categories: (1) the pie-slice  $f$ - $k$  (frequency-wavenumber) filters [25]; (2) the  $\tau$ - $p$  (intercept-time, slope/slowness) transform, also called Radon or slant-stack transform [20, 25]; (3) the  $p$ - $f$  (slowness-frequency) filters [12, 7]; and (4), the filters based on eigenvalue decompositions [23]. Note that, except the last one, all these techniques are closely related; the  $f$ - $k$  and  $\tau$ - $p$  filters by the projection slice theorem [9], and the  $p$ - $f$  and  $\tau$ - $p$  transforms by the Fourier transform.

These techniques are frequently adapted to be used in a local fashion to be able to follow the slowness, amplitude and waveform variations that seismic waves usually exhibit at the cost of losing slowness resolution. The main features that distinguish the different approaches available are: the redundancy factor and the slowness/directional resolution. Directional filter-banks [1, 21] used for image processing are usually designed with reduced redundancy but mid-low slowness resolution; while continuous approaches, such as local slant-stack transform (LSST) [13, 3] or combinations between Radon and wavelet transforms [16, 18, 15], yield mid-high slowness resolution and high redundancy. Common applications on seismic signals processing are noise filtering [14] and migration operations [8].

In this paper we compare two complementary approaches: (1) a local time-scale slant-stack transform, highly redundant but with a high slowness resolution and (2) a directional multiscale wavelet transform based on dual-tree  $M$ -band filter-banks with a two-fold redundancy but with a low slowness resolution. The organization of the paper is as follows: Section 2 describes the proposed LSST with an analysis and a combined filtering/synthesis approach. Section 3 briefly recalls the principles behind the  $M$ -band dual-tree wavelets. Comparisons between the two methods are drawn in Section 4 on a real seismic dataset, followed by conclusive comments.

## 2. LOCAL ADAPTIVE SLANT STACK FILTERS IN THE TIME-SCALE DOMAIN

Broadly, an improvement of the accuracy of seismic signal waveform estimators in the LSST domain implies an increment of the number of samples used along the time or the

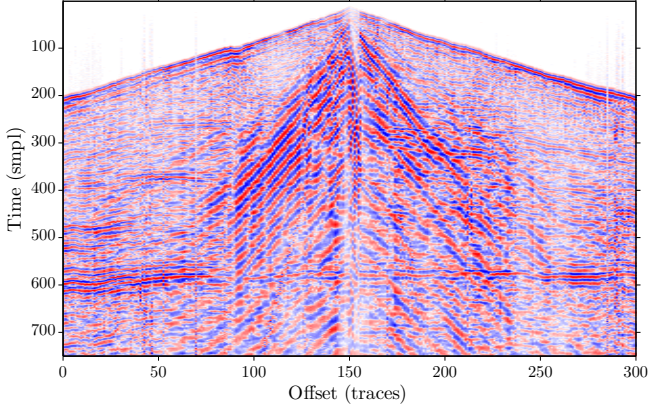


Figure 1: Original noisy seismic data in shot gather.

offset axis. However, the instantaneous slowness variations of the seismic signals and, to a lesser extent, their amplitude variations, extremely limits the number of samples allowed along the offset axis. Moreover, we can not perform any substantial smoothing in the time axis; even though the instantaneous slowness of coherent signals are slowly varying in time, the seismic waveform is measured from its instantaneous amplitude, which varies fast.

The direct use of the instantaneous amplitude in the waveform estimation entails the application of the same filter for all the frequency/scales components of the signal. Meanwhile, the rich spectral content of the seismic signals makes much more desirable to be able to adapt the filter at each frequency/scale.

In the proposed approach, we decompose each seismic trace in slowness in a time-scale domain to increase the signal to noise ratio and the seismic event tracking capability with respect an adaptive LSST approach, while keeping the time resolution. The high degree of freedom that the time-scale domain provides enables the design of a large set of filters much more selective in slowness than the equivalent LSST ones. This freedom allows us to set the optimum slowness resolution at each frequency, or in opposition, to keep the resolution constant to build filters that do not modify the waveform of the seismic signals processed. Additionally, it is also possible to configure these filters to mimic a large variety of slowness filters in the  $f$ - $k$ ,  $p$ - $f$  and  $\tau$ - $p$  domain, reducing problems of aliasing in the first ones and of inversion in the last ones.

## 2.1 Analysis

The algorithm performed by the local time-scale slant-stack transform is a combination of the LSST and a continuous wavelet transform [11]. The LSST of a seismic profile  $u$  with a nonuniform separation between traces can be written as locally weighted sums of  $L$  neighboring traces along a set of signal trajectories of slope/slowness  $p_s$ , being  $s$  the slope/slowness index:

$$v_{s,m}[n] = \sum_{l=-(L-1)/2}^{(L-1)/2} g_m[l] u_{m+l}(nT + (d_{m+l} - d_m) p_s) \quad (1)$$

where each element  $v_{s,m}[n]$  of the LSST decomposition is an estimation of the contribution of the signal  $s$  to the sam-

ple  $u_m(nT)$ , being  $n$  and  $m$  the sample and trace indices, respectively, and  $T$  the sampling period. The delay term  $(d_{m+l} - d_m) p_s$  models the time that the signal  $s$  needs to travel the distance from the offset  $d_m$  to  $d_{m+l}$ . The space window  $g_m[l]$  may depend on the offset of the  $m$  trace, due to the usually non uniform separation between traces. And  $L$  may change along any dimension.

The equivalent operation in the time-scale domain can be written as the above LSST now performed on the time-scale domain:

$$Wv_{s,m}[n, j] = \sum_{l=-(L-1)/2}^{(L-1)/2} g_\lambda[l] W u_{m+l}(nT + (d_{m+l} - d_m) p_s, 2^j) \quad (2)$$

where  $W u_m(\tau, \lambda)$  stands for the wavelet transform along time of the seismic profile  $u(t, x)$  and each element  $W v_{s,m}[n, j]$  is the contribution of the scale  $\lambda = 2^j$  of the signal  $s$  at the sample  $n$  of the trace with offset  $x = d_m$ . As before,  $L$  may change along any dimension.

For a proper performance of the above slowness decomposition, the wavelet transform and its inverse have to fulfill three essential features: nearly perfect reconstruction, to reduce estimation errors on the slowness components of the signal; linear phase delay, to preserve the seismic waveform; and close-to time-invariance, to reduce the interpolation error at required time-scale positions not provided by the discretized wavelet transform. Additionally, it is useful to have freedom in the choice of the mother wavelet. For this reason, we have chosen an oversampled complex wavelet transform based on frames of wavelets with several voices per octave.

## 2.2 Filtering and synthesis

A seismic section decomposed in time, space, scale and slowness that contains  $R$  coherent signals of  $q_r$  slowness,  $u_r(t - q_r x)$  with  $r \in [1, R]$ , can be approximately modeled as:

$$Wv_{s,x_c}(\tau, \lambda) \simeq \sum_{r=1}^R h(\lambda, q_r - p_s) W u_r(\tau - q_r x_c, \lambda) \quad (3)$$

where  $h(\lambda, q_r - p_s)$  denotes a transfer function that models the cross-interference in the slowness axis.

This transfer function depends on the LSST space window  $g(x)$ , and the mother wavelet.

$$h(\lambda, q_r - p_s) = \frac{1}{\lambda^2 C_\Psi} \int_{-\infty}^{\infty} \Psi_\lambda(\tau) \Psi_{1_\lambda}^*(\tau, q_r - p_s) d\tau \quad (4)$$

where  $\Psi_{1_\lambda, s}(t)$  is the family of functions used in the time-scale LSST,

$$\Psi_{1_\lambda}(\tau, q_r - p_s) = \int_{-\infty}^{\infty} \frac{1}{|q_r - p_s|} g_\lambda\left(\frac{t}{q_r - p_s}\right) \Psi_\lambda(t - \tau) dt \quad (5)$$

In this work, we synthesized the filtered section using a conventional 1-D inverse continuous wavelet transform, but other options are possible, such as a unique family of functions for analysis and synthesis, similar to the approach of curvelet frames [15].

The optimum scaling of  $g(x)$  is problem dependent, and generally not linear with the scale. As a consequence,

$\psi_{1,\lambda,s}(t)$  can not be obtained by scaling and translating a unique mother wavelet; except on the particular case where the window length is proportional with the scale, where the transfer function is scale independent,

$$h(q_r - p_s) = \frac{1}{C_\psi} \int_0^\infty \hat{g}((q_r - p_s)\omega) |\hat{\psi}(\omega)|^2 \frac{d\omega}{\omega} \quad (6)$$

and the model (3) is exact. In Eq. 6,  $\hat{\psi}(\omega)$  and  $\hat{g}(\omega)$  denotes the Fourier transform of the 1-D mother wavelet and the LSST space window, respectively.

The time-scale LSST based filter can be represented as the weighted sum of the decomposed signal,

$$W_{y_{x_c}}(\tau, \lambda) = \sum_{s=1}^S \sum_{r=1}^R f(\tau, \lambda, p_s) h(\lambda, q_r - p_s) W_{u_r}(\tau - q_r x_c, \lambda) \quad (7)$$

where  $W_{y_{x_c}}(\tau, \lambda)$  is the filtered seismic signal in the time-space-scale domain and  $f(\tau, \lambda, p_s)$  the filter to design.

If we define  $k_r(\tau, \lambda)$  as the desired gain for the  $r$  signal, the filter to design has to satisfy as many equations as coherent signals exist. From (7),

$$\sum_{s=1}^S f(\tau, \lambda, p_s) h(\lambda, q_r - p_s) = k_r(\tau, \lambda) \quad r \in [1, R] \quad (8)$$

or in vector notation,

$$\mathbf{H}(\lambda) \mathbf{f}(\tau, \lambda) = \mathbf{k}(\tau, \lambda) \quad (9)$$

This system of equations is underdetermined,  $S \gg R$ , and it thus possesses an infinite number of solutions. From all these solutions, in the particular case that the instantaneous slowness of all the coherent signals are known, the least square solution leads to the minimum level of noise in the slowness axis. Considering a white Gaussian noise, the solution of (9) under the minimum 2-norm constraint:

$$\min_{\mathbf{f}(\tau, \lambda)} \mathbf{f}^T(\tau, \lambda) \mathbf{f}(\tau, \lambda) \quad \text{that} \quad \mathbf{H}(\lambda) \mathbf{f}(\tau, \lambda) = \mathbf{k}(\tau, \lambda) \quad (10)$$

is

$$\mathbf{f}(\tau, \lambda) = \mathbf{H}(\lambda) (\mathbf{H}^T(\lambda) \mathbf{H}(\lambda))^{-1} \mathbf{k}(\tau, \lambda) \quad (11)$$

To design this 4-D filter it is necessary to set  $k_r(\tau, \lambda)$  appropriately. With the aim of designing this filter automatically, we estimated the local maxima of the modulus of the complex time-scale LSST through instantaneous slowness measures of the most coherent signals in the slowness range of interest.

### 3. M-BAND WAVELETS AND HILBERT PAIRS

Let  $M$  be an integer greater than or equal to 2. An  $M$ -band multiresolution analysis of  $\mathbb{L}^2(\mathbb{R})$  is defined by one scaling function  $\psi_0 \in \mathbb{L}^2(\mathbb{R})$  and  $(M-1)$  mother wavelets  $\psi_m \in \mathbb{L}^2(\mathbb{R})$ ,  $m \in \{1, \dots, M-1\}$  [19], solutions of the following scaling equations:

$$\forall m \in \{0, \dots, M-1\}, \quad \frac{1}{\sqrt{M}} \psi_m\left(\frac{t}{M}\right) = \sum_{k=-\infty}^{\infty} h_m[k] \psi_0(t-k), \quad (12)$$

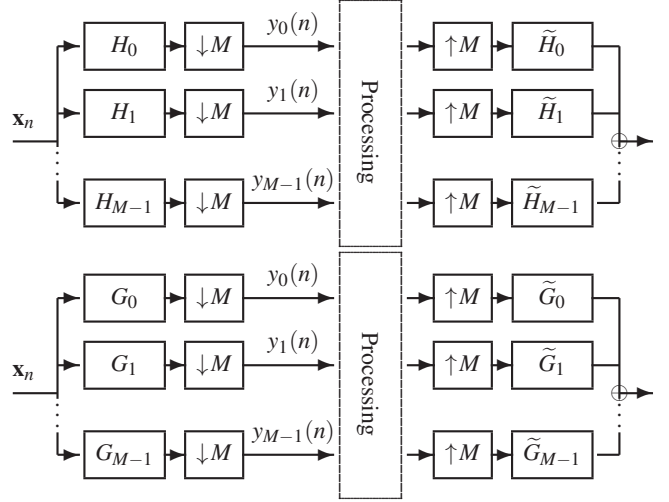


Figure 3: A pair of analysis/synthesis  $M$ -band para-unitary filter banks generating  $M$ -band dual-tree wavelets.

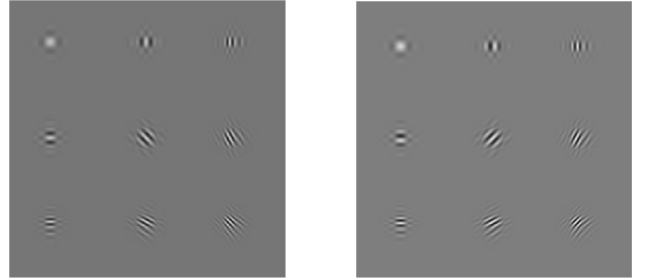


Figure 4: Bidimensional 3-band dual-tree wavelets: (Left) negatively oriented scaling function  $\psi_0$  and 8 wavelets  $\psi_k$ ; (Right) positively oriented scaling function  $\psi_0^H$  and 8 wavelets  $\psi_k^H$ .

where the sequences  $(h_m[k])_{k \in \mathbb{Z}}$  are square integrable. A “dual”  $M$ -band multiresolution analysis is defined by a scaling function  $\psi_0^H$  and mother wavelets  $\psi_m^H$ ,  $m \in \{1, \dots, M-1\}$  related to the functions  $\psi_m$  by an Hilbert pair relationship. More precisely, the dual mother wavelets will be obtained by an Hilbert transform from the “primal” wavelets  $\psi_m$ ,  $m \in \{1, \dots, M-1\}$ . In the Fourier domain, the Hilbert transform reads:

$$\forall m \in \{1, \dots, M-1\}, \quad \hat{\psi}_m^H(\omega) = -i \text{sign}(\omega) \hat{\psi}_m(\omega). \quad (13)$$

where the signum function sign is defined as:

$$\text{sign}(\omega) = \begin{cases} 1 & \text{if } \omega > 0 \\ 0 & \text{if } \omega = 0 \\ -1 & \text{if } \omega < 0. \end{cases} \quad (14)$$

These dual wavelets also satisfy two-scale equations similar to Eq. (12) with the square integrable sequences  $(g_m[k])_{k \in \mathbb{Z}}$ . The general representation in terms of filter banks, on one decomposition level, is shown in Fig. 3. Figure 4 represents the 2-D basis functions obtained [4] with  $M = 3$  bands.

They illustrate that different directions can be extracted

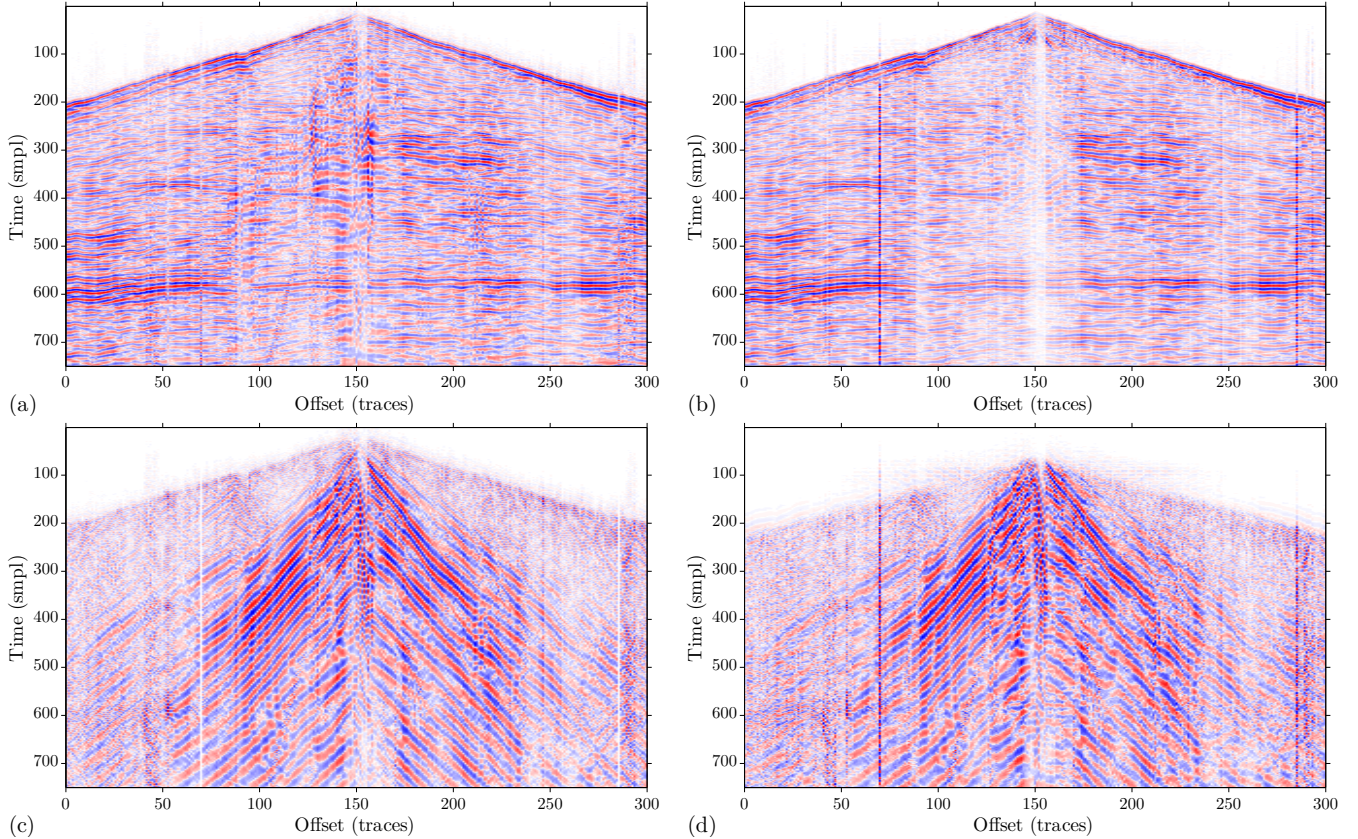


Figure 2: Filtered signal (a) using the time-scale LSST to remove the high slope interferent signals, and (b) with local mute of dual-tree coefficients, together with their differences w.r.t. original data in (c) and (d).

from the transform, since band- and high-pass wavelets select opposite directions for each tree. Since the present paper focuses on applications, we refer to [17, 5] for further details on dual-tree wavelet transforms. In contrast to traditional wavelet shrinkage where small coefficients are discarded, high-valued coefficients are masked since they correspond to aforementioned coherent noises [10].

#### 4. RESULTS AND DISCUSSION

To compare the two approaches, we have applied these techniques to remove coherent noise in the dataset in Figure 1. It represents  $750 \times 300$  seismic image acquired in a rugged tomography foothill province. The occluding coherent noise under consideration is located at the acute cone originating from the central apex. Its highly energetic and directional characteristics, since it does not bear significant information, prevents the estimation of geological substructures with lower amplitude.

To better qualify the filters' behaviour and results, the Structural Similarity Index (SSIM) [24], now commonly employed for image quality assessment, has been applied. The SSIM was initially proposed to overcome the limitations of standard metrics used in image comparison and evaluation such as mean squared error or peak signal-to-noise ratio. Although the nature of seismic data is rather specific and differs from natural images, computing similarity maps (Fig. 5) between the original data and filtered estimations provides meaningful insights. Local high SSIM measures between

original and filtered data correspond to close shape.

Most of the differences between the results obtained with both filtering techniques, apparent on Figures 2 and 5, are related to their different space-slowness trade-off. Both techniques attenuate the main high-slope coherent-noise components (mid part on all these figures) with a higher attenuation in the case of the dual-tree based filter — high SSIM value — thanks to its shorter filter response or equivalently higher space resolution w.r.t. the other. However, due to its lower slowness resolution the distortion on low-slope signals are higher. This effect can be noticed on the higher SSIM measures on the outer part of Fig. 5(b), while the measures of Fig. 5(a) are lower since the time-scale LSST based filter gives a higher slowness resolution in accordance with its longer filter response. High-resolution figures are made available at [22].

#### 5. CONCLUSIONS

Time-scale directional filters are a powerful tool that can significantly improve seismic data processing, thanks to the enhancement in the seismic wave detection, separation and tracking capabilities. The flexibility of these tools enables adaptive instantaneous slowness with a fair degree of control of the slowness resolution in time-scale, while keeping a minimum level of distortion to the signals under analysis. As interferences are processed at each trace in the time-scale-slowness domain, it becomes possible to isolate them, in contrast to the commonly used  $\tau$ - $p$  transform and  $f$ - $k$  pie-

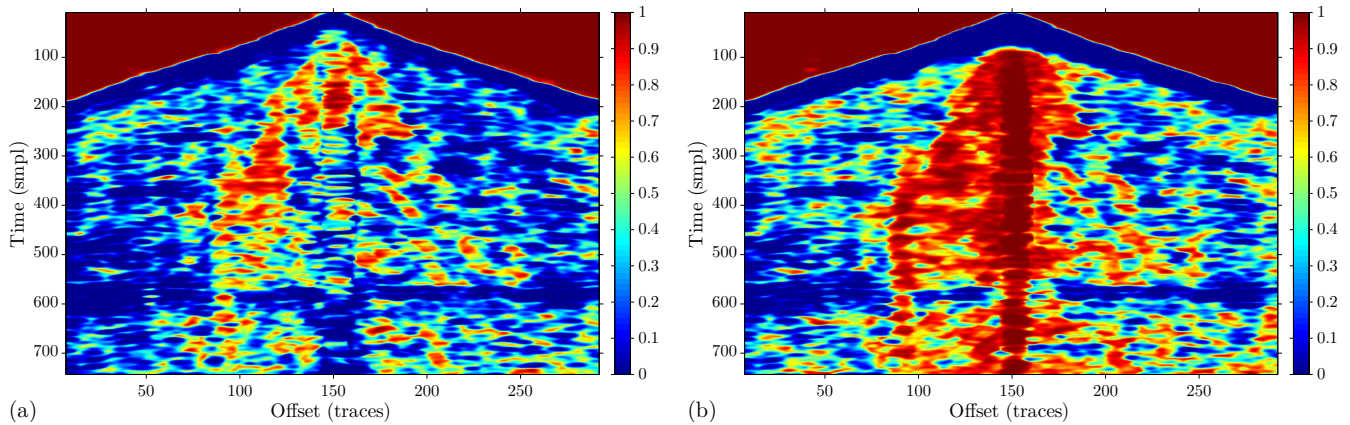


Figure 5: Similarity maps between original data and estimated interferent signals using (a) the time-scale LSST, and (b) the dual-tree wavelets.

slice based filters because of the lack of space resolution of the former and of time-space resolution of the latter. The two techniques used in this context yield two complementary approaches, with a different balance in slowness resolution and redundancy. Both are able to attenuate the main part of the chevron-like coherent noise occluding meaningful geologic information. The time-scale LSST focuses on a reduced distortion level of the signal of interest, while the dual-tree wavelet transform aims at increased noise rejection. Due to the similarity of the time-scale LSST with other directional transforms used in image processing [2, 15], the quest for a vaster family of adaptive transforms linking proposed methods deserves further investigations.

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