## The Singular Spectrum Analysis method and its application to seismic data denoising and reconstruction

by

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Department of Physics

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## Abstract

Attenuating random and coherent noise is an important part of seismic data processing. Successful removal results in an enhanced image of the subsurface geology, which facilitate economical decisions in hydrocarbon exploration. This motivates the search for new and more efficient techniques for noise removal. The main goal of this thesis is to present an overview of the Singular Spectrum Analysis (SSA) technique, studying its potential application to seismic data processing.

An overview of the application of SSA for time series analysis is presented. Subsequently, its applications for random and coherenet noise attenuation, expansion to multiple dimensions, and for the recovery of unrecorded seismograms are described. To improve the performance of SSA, a faster implementation via a randomized singular value decomposition is proposed. Results obtained in this work show that SSA is a versatile method for both random and coherent noise attenuation, as well as for the recovery of missing traces.

To my family...

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# List of symbols and abbreviations

$[]^H$	Hermitian or conjugate transpose of a matrix, page 10
$[]^T$	Transpose of a matrix, page 12
A	Average on the anti-diagonals operator, page 14
M	Hankelization operator, page 12
λ	Discrete eigen-value of a squared matrix, page 9
$\mathcal{O}$	Order of magnitude, page 46
ω	Temporal frequency, page 22
$\sigma$	Discrete singular-value of a matrix, page 13
i	Imaginary unit $i = \sqrt{-1}$ , page 22
f - x	Frequency-Space domain, page 3
$f_{final}$	Final frequency to analyze in the SSA process, page 53
$f_{init}$	Initial frequency to analyze in the SSA process, page 53
i	Power iteration variable set by the user in the randomized algorithm for rank reduction, page $46$
K	Number of lagged vectors in a Hankel matrix, page 12
k	Final rank of a matrix (related to the number of events in a seismic record), page $13$
L	Length of the lagged vectors, page 12
l	Oversampling integer for the randomized rank reduction algorithm, page $45$
$l_i$	Lagged vectors from a time series, page 12
$L_x$	Length of the lagged vectors in the $x$ dimension. This parameter also controls the dimensions of the Hankel matrix, page 24
m	Number of columns of a matrix, page 10

- n Number of rows of a matrix, page 10
- $N_x$  Number of space samples in the data, page 24
- p Ray parameter representing the dip of a 2-D waveform, page 22
- s(t) 1-D time series, page 11
- $S(x,\omega)$  2D waveform in the f-x domain, page 22
- s(x,t) 2D waveform with constant dip, page 21
- $S(x, y, \omega)$  3-D waveform in the f x domain, page 29
- s(x, y, t) 3-D waveform with constant dip, page 29
- $s_k$  Time series recovered from the rank reduced Hankel matrix by averaging along the anti-diagonals, page 14
- t time, page 22
- t-x Time-Space domain, page 3
- w(t) Pulse or Wavelet, page 22
- x Space in the x dimension, page 22
- y Space in the y dimension, page 29
- $\Lambda$  Diagonal matrix containing the eigen-values of a square matrix, page 10
- $\Sigma$  Diagonal matrix containing the singular values of a matrix, organized in descending order, page 10
- $\Sigma_k$  Matrix presenting zeros in all but the first k elements of the diagonal matrix  $\Sigma$ , page 11
- **C** Example of a Hermitian matrix, page 10
- M Hankel Matrix, page 12
- $\mathbf{M}_k$  Rank reduced Hankel matrix to rank k, page 14
- $\mathbf{S}_{\omega}$  Spatial array of a given frequency  $\omega$ , page 24
- **U** Squared unitary matrix presenting the eigen-vectors of  $\mathbf{X}\mathbf{X}^{H}$ , page 10
- $\mathbf{U}_k$  Matrix presenting the first k columns of U, page 11
- $\mathbf{U}_k \mathbf{U}_k^H$  Rank reduction operator, page 11
- **V** Squared unitary matrix presenting the eigen-vectors of  $\mathbf{X}^{H}\mathbf{X}$ , page 10
- $\mathbf{V}_k$  Matrix presenting the first k columns of  $\mathbf{V}$ , page 11
- $\mathbf{W}_{i}(x)$  Waveform that represents each individual event *i* in a seismic record, page 63

- X General matrix representing a group of data measurements, page 9
- I Matrix presenting ones in all its cells, page 53
- T Operator that identifies the presence of traces on the data, page 53
- $\widehat{\mathbf{S}}_{\omega}$  Filtered version of  $\mathbf{S}_{\omega}$  after averaging over the anti-diagonals, page 26
- $\widetilde{\mathbf{M}}_k$  Randomized approximation to the rank k Hankel matrix  $\mathbf{M}_k$ , page 45
- CDP Common Depth Point, page 6
- CMP Common Midpoint, page 1
- GR Ground Roll, page 61
- LMO Linear Move-out, page 70
- MSSA Multichannel Singular Spectrum Analysis, page 5
- NMO Normal Move-out, page 1
- POCS Projection onto convex sets, page 6
- R-SVD Randomized SVD, page 6
- SSA Singular Spectrum Analysis, page 3
- SVD Singular Value Decomposition, page 4

## CHAPTER 1

## Introduction

#### 1.1 Background

The seismic method is an important geophysical tool for the study of subsurface geology. It allows one to obtain geological information over an extensive area without having to measure its properties directly. The seismic method consists of generating a wavefront that propagates through the ground that is then recorded at the surface using an array of receivers. Although most of the energy of this wave is absorbed by the earth, some energy is reflected by subsurface structures and recorded at the surface by arrays of receivers. Ground displacements produced by different waves, including these reflections, are recorded by the receivers and saved into seismograms. In a typical land seismogram one can identify several waveforms, such as a direct wave, refractions, reflections and ground roll (Figure 1.1). Since the reflections travel deeper into the subsurface than other types of wave, they are generally the main target in seismic surveys. Thus, seismograms are often processed and inverted to enhance the signal from the reflections, producing a cleaner image of the subsurface.

A basic processing sequence starts with preprocessing and deconvolution techniques, followed by common midpoint (CMP) sorting, velocity analysis, normal move-out (NMO) correction and stacking. After a preliminary image is obtained it is improved by applying a residual statics correction, poststack processing and migration. Preprocessing consists of correcting the elevation statics and in filtering those elements of the records that interfere with the reflections. Deconvolution aims to increase the frequency band of the signal. In addition, CMP sorting consists of a reorganization of the traces, grouping together those that have the same geographical midpoint between source and receiver. Velocity analysis and NMO correction are steps that analyze the arrival times of the reflections and performs a correction that horizontalizes them. Then, the stacking step proceeds to average those traces in each CMP. The next steps are the residual statics corrections and poststack processing, which objectives are to filter the noise that was not removed in the process and to improve the lateral continuity of the events. Finally, migration applies inversion techniques to recover the true position of the events. The result of this entire process is an image of the subsurface that can be used to create a model of the geological features in the area. It is clear that each stage depends on the results of previous steps. Therefore, noise attenuation methods applied in preprocessing steps of the sequence are fundamental to obtain results of high quality. Failure in the aforementioned processes may result in a seismic image of low quality. This ultimately leads to increased difficulties when making economical and logistical decisions pertraining the development of an exploration play.



Figure 1.1: Features of a seismic record.

There are different types of noise that can be found in seismic data. According to their behavior they can be classified into coherent and incoherent noise. Coherent noise is present in adjacent traces, and is generally generated by the source or by the interaction of the main wavefield with the ground. Some forms of coherent noise are multiple reflections, ghosts, ground roll, air wave, etc. The incoherent noise, also called random noise, is recorded by each receiver independently, meaning it does not correlate between adjacent channels. The latter can be produced by environmental factors, such as the wind moving vegetation in the survey area. The vibrations produced by the external factors are recorded by nearby receivers. Given that the energy of these waves is very small, they are only detected by nearby channels, making this noise incoherent. Any other perturbation in the surroundings of a receiver can also generate random noise.

The main objective of this thesis is to study the application of a rank reduction method, called Singular Spectrum Analysis (SSA), for the attenuation of random noise and ground roll. The results obtained from the application of this method show a significant improvement in the noise reduction compared with traditional techniques. In the next section I will review some of the traditional techniques for noise reduction, which leads to the motivation of this work.

#### **1.2** Noise attenuation methods

The attenuation of random and coherent noise in seismic records is an important subject in seismic data processing. In general, random noise is attenuated in the CMP stacking step of processing, but in many cases noise is not completely removed by stacking. Classical methods for random noise attenuation exploit the predictability of the seismic signal in small spatio-temporal windows. An example of the aforementioned concepts is the f - x deconvolution (Canales, 1984), which takes advantage of the properties of the signal in the f - xdomain. In this domain, the signal is predictable as a function of the space. Random noise is attenuated by applying a complex Wiener prediction filter that exploits this predictability (Gulunay, 1986). Variations of f - x deconvolution are focused in improving the design of the filter. For example, Sacchi and Kuehl (2001) introduced an autoregressive/movingaverage (ARMA) model to represent the signal. Nevertheless, those methods take advantage of the same properties of the signal in the f - x domain. Another example of a method that exploit the predictability of the signal is the t-x prediction error filter (Abma and Claerbout, 1995), which works in the time-space domain by applying a single prediction filter using a conjugate gradient method. Although it is important for noise reduction techniques to be able to significantly attenuate the noise, it is also important to produce outputs with minimal signal distortion. This condition is only met in f - x deconvolution for low and medium levels of noise. The signal distortion can be high in low signal-to-noise-ratio situations (Harris and White, 1997; Gulunay, 2000).

Another category of methods rely on rank reduction techniques to decompose a window of seismic data in coherent and incoherent components (Ulrych et al., 1999). Examples in this

category are abundant in the geophysical literature. Freire and Ulrych (1988), for instance, proposed to carry out rank reduction of seismic images in the t - x domain via the socalled eigen-image decomposition. This approach takes advantage of the linear dependency between traces, so it works well for horizontal events. Chiu and Howell (2008) and Cary and Zhang (2009) extended this idea for the elimination of ground roll. For this purpose the offending event (ground roll) is flattened via a linear moveout (LMO) correction, and then modelled using the eigen-image method to be finally subtracted from the initial data. The eigen-image method is also connected to the Karhunen-Loéve transform (KL) and principal component analysis (PCA) methods (Freire and Ulrych, 1988). The KL transform has been applied by Jones and Levy (1987), Marchisio et al. (1988) and Al-Yahya (1991) for signalto-noise ratio enhancement in seismic records. KL transform and PCA are similar methods that use Singular Value Decomposition (SVD) for their applications and are sometimes used as equivalent. The main differences between these two methods are compiled by Gerbrands (1981). In general, rank reduction methods that are applied in the time-space domain, are unsuccessful in identifying dipping events.

A rank reduction method that is independent of dip, and therefore, does not require flattening, has been proposed by Mari and Glangeaud (1990). This method is called Spectral Matrix Filtering; it was presented as an alternative to separate up-going and down-going waves in VSP records. This method operates in the f - x domain and requires the eigendecomposition of the spectral matrix of the data. These techniques have been expanded also to several dimensions of seismic data. Trickett (2003) proposed the application of an eigen-image filter that works in the f - xy domain, reducing the rank of the spatial matrix in each frequency. The latter is called f - xy eigenimage noise suppression. Although this filter performs in multiple dimensions, the improvement of the signal-to-noise ratio of the results is low compared to other techniques. Other rank reduction methods for noise filtering apply a reorganization of the rows or columns of the data matrix to improve the coherency of the signal. One of these methods is the truncated SVD, which works in time slices of a stacked data cube by rearranging its columns into a Hankel matrix to suppress acquisition footprints and random noise on stacked data (Al-Bannagi et al., 2005).

Although rank reduction methods have been used in seismic data processing for many years, there is a method that has been recently attracting attention. This method is SSA (Vautard et al., 1992), which is the main topic of this thesis. SSA arises from the decomposition of time series in the study of dynamical systems (Broomhead and King, 1986). It has been well studied in many fields, like climatic series analysis (Vautard and Ghil, 1989; Ghil et al., 2002), astronomy (Auvergne, 1988; Varadi et al., 1999) and medicine (Mineva and Popivanov, 1996; Aydin et al., 2009), but it is subjected to ongoing research in seismic data processing. SSA works in the f - x domain and consists of reorganizing spatial data into a Hankel matrix. Reducing the rank of this Hankel matrix can reduce the random noise in the record without distorting the signal. The latter provides a significant advantage over traditional noise attenuation techniques like f - x deconvolution. The SSA method has also been called Cadzow filtering (Trickett, 2008) or the Caterpillar method (Golyandina et al., 2001). All this techniques are equivalent, but they arise from different fields. For instance, the Cadzow method was proposed as a general framework for denoising images (Cadzow, 1988), and the Caterpillar method also arises from time series analysis (Nekrutkin, 1996). Research in the application of this method for random noise attenuation has been published by Trickett (2008), Sacchi (2009) and Trickett and Burroughs (2009).

This thesis presents an overview of SSA, beginning with an explanation of its origins for time series analysis. The application of SSA for random noise attenuation is then explained in detail, including its expansion to multiple dimensions. A main drawback of SSA is that it can be significantly slow compared to other seismic attenuation methods. To solve this, I propose the introduction of a randomized algorithm for rank reduction, developed by Rokhlin et al. (2009), which decreases the amount of computations required by SSA. In addition to random noise attenuation, I will also investigate two different applications for SSA. In particular, an iterative algorithm to recover missing traces in an irregularly sampled data set is proposed. Another application is the use of SSA to attenuate ground roll, which takes advantage of the capacity of SSA to separate different events. With this work I aim to generate a compilation of applications of SSA for seismic data processing, which can be used as a base to more specialized seismic data processing studies.

#### **1.3** Organization of the thesis

This thesis is organized as follows:

- Chapter 2 expands on the origins of SSA as a time series analysis technique. It explains the steps for the application of SSA in the analysis of dynamical components of a time series, signal detrending and noise attenuation. This chapter presents an example of the application of SSA for noise attenuation, by recovering a sinusoidal curve contaminated with noise. It also shows the application of SSA to decompose the Wolf sunspots number curve into its singular spectrum, which can provide information of the processes that control the data. The interpretation of these components in not discussed here, limiting the explanation to the application of SSA.
- Chapter 3 shows the application of SSA for random noise attenuation of seismic records. This chapter also introduces the expansion of SSA to multiple dimensions called Multichannel Singular Spectrum Analysis (MSSA). This expansion is explained for 2-D MSSA and N-D MSSA. Several examples of SSA are presented where random

noise is attenuated. Examples include synthetic gathers with linear and hyperbolic events, as well as real post-stack gathers. Examples of MSSA are limited to the 2-Dimensional case, in which the random noise of a synthetic cube with linear events is attenuated.

- Chapter 4 presents the application of a new rank reduction algorithm to the SSA technique. This algorithm is a randomized SVD (R-SVD) that generates an approximation to the rank reduced matrix required by SSA. The randomized algorithm requires a significantly lower amount of calculations compared with the traditional SVD algorithm. Examples in this chapter show the results from the application of MSSA using SVD and R-SVD in the rank reduction step. The amount of time required by each process is also shown. The results of this test show that the application of the R-SVD algorithm in MSSA decreases by 50% the running time of the method.
- Chapter 5 explores the use of SSA for seismic data interpolation. The algorithm of MSSA is changed to work recursively, performing several iterations. In each iteration the missing traces of the input data are replaced by the traces recovered by MSSA. After a limited number of iterations the signal is reconstructed. This iterative algorithm is similar to an algorithm called Projection onto convex sets (POCS). The example in this chapter consists on a real Common Depth point (CDP) gather which offsets are not regular. The data are regularized into a desired grid and the cells with missing traces are recovered using the iterative MSSA algorithm. Although this is the only example of MSSA applied to real data, it shows how the application of MSSA can recover missing data and, at the same time, can attenuate random noise.
- Chapter 6 presents a different approach for the application of SSA. It expands on the use of SSA for ground roll attenuation. The principles of this technique are based on the property of SSA to identify different linear events when one of the first singular values is recovered independently. This separation can only take place under certain conditions, which are explained in detail. A synthetic example shows the separation of two events, one of which presents a significantly lower frequency and different velocity than the other. The difference in frequency and velocity simulates ground roll and reflections. A second example applies this technique to a real shot gather that presents a strong ground roll. The results show that signal separation using SSA is successful in attenuating ground roll having low effect on the reflections.
- Chapter 7 presents the conclusions and recommendations for further work.

## CHAPTER 2

# Singular Spectrum Analysis in the study of time series

#### 2.1 Background

SSA is a model free technique that arises from the research of alternative tools for 1-D time series analysis. It results from the analysis of the singular spectrum of a trajectory matrix constructed from the time series of interest. Early applications of SSA are focused on the analysis of dynamical systems. It is used to identify degrees of freedom in time series, and this way, find the main physical processes present in the data. An important contribution to the development of SSA was made by Broomhead and King (1986), who used the method of delays proposed by Takens (1981) to study dynamical systems using multivariate statistical analysis. Independently, Fraedrich (1986) also applied SSA to the dimensional analysis of paleoclimatic marine records. SSA is studied more in depth by Vautard et al. (1992), who presents it as a tool for the analysis of short, noisy and chaotic signals. They investigate four major problems that arise from the application of SSA. These problems are: how parameters like the embedding dimension influences the analysis, what is the level of robustness and statistical confidence of the results, the possible applications in the identification of noise and how to interpret the information given by each singular component. The basic aspects of SSA are compiled and explained in books from Elsner and Tsonis (1996) and Golyandina et al. (2001), which complement the information with different examples and applications.

SSA is a common tool in climatic series analysis. Vautard and Ghil (1989) and Yiou et al. (1996) used this technique to study the main oscillations in paleoclimatic records, identifying the amount of degrees of freedom in the data. It was also used to study baroclinic

processes (Read, 1992). Even though SSA has been mostly applied in meteorological studies, other disciplines have found it useful. In astronomy, for example, it have been applied for phase space reconstruction of pulsating stars (Auvergne, 1988) and for the detection of low-amplitude solar oscillations (Varadi et al., 1999). In medicine, it has proven useful in decomposing data from electroencephalograms, to analyze the preparation time before a voluntary movement (Mineva and Popivanov, 1996) or to support clinical findings in insomnia (Aydin et al., 2009). It has even been used for time series forecast by Danilov (1997) and Golyandina et al. (2001). In economy, it has been used for the analysis and forecasting of time series like daily exchange rate (Hassani et al., 2010) or the agricultural crop yield, milk production and purchase, number of road traffic accidents, etc (Polukoshko and Hofmanis, 2009).

The application of SSA can be expanded to multiple time series simultaneously. This is called multivariate or multichannel singular spectrum analysis (MSSA) which was first applied by Read (1993). The difference with SSA is that the trajectory matrix includes information from all the time series analyzed. In his study, Read (1993) applies MSSA to phase analysis of time dependent experimental temperature measurements taken simultaneously. Again, the main application for this technique is to study climatic records. Plaut and Vautard (1994) uses this technique to study climatic low frequency oscillations in mid latitudes on the northern hemisphere. It is applied to study the variations in the tropical Pacific climate (Hsieh and Wu, 2002) and is included in a review of the application of spectral methods to climatic data presented by Ghil et al. (2002). In a different approach, MSSA is also applied to signal reconstruction and forecasting of time series (Golyandina and Stepanov, 2005) and for the filtering of digital terrain models (Golyandina et al., 2007).

Together with the application of SSA in the study of dynamical systems, it is found useful for noise attenuation in time series. This is carried on by recovering an inferior number of singular values after the decomposition of the data. This property was observed by Broomhead and King (1986) and is implemented in many studies. It has been said that SSA is more powerful as a denoising technique than as a tool for dynamical analysis (Mees et al., 1987; Paluš and Dvořák, 1987). The main challenge in the use of SSA for noise attenuation arises from the selection of the number of singular values that recover the data. The answer to this depends on how correlated are the signal and the noise. In general, the signal is believed to be represented by the largest singular values (Elsner and Tsonis, 1996), but this can change when the noise is not white or the signal-to-noise ratio is too low. Most of the papers that treat the application of SSA to dynamical systems also expand on its application for noise removal. Works that investigate less subjective ways to use SSA for the attenuation of noise in time series are carried on by Hansen and Jensen (1987); Allen and Smith (1997) and Varadi et al. (1999). The application of SSA consists of four main steps. The first step is the embedding of the time series, which consists of organizing its entries in a trajectory matrix. The second step is the decomposition of this matrix in its singular spectrum by using SVD. The third step consists of the application of a rank reduction to the trajectory matrix by recovering fewer amounts of singular values from the decomposition. Finally, the time series is recovered from the rank reduced trajectory matrix. This chapter presents the basic theory behind SSA, emphasizing in its application for decomposing and denoising time series. Further details on the four main steps for the application of SSA are presented, together with examples that demonstrate its effectiveness in extracting main oscillations and increasing the signal-to-noise ratio of the data. A particular approach to MSSA will be done in Chapter 3, where it will be used for the attenuation of random noise in seismic records.

#### 2.2 Preliminaries

Many rank reduction methods rely on the application of the SVD technique to calculate the rank reduced matrix. Before explaining the application of SSA to time series and seismic record denoising it is necessary to understand the rank reduction process. In this section the process of rank reduction using SVD will be expanded.

A group of data measurements can be viewed as a matrix  $\mathbf{X}$ . For example, 1-D time series can be represented as a matrix  $\mathbf{X}$  by computing its autocorrelation matrix or by embedding its components. In seismic surveys, the columns of a seismic section may represent traces and the rows time samples of each trace. Seismic data in other domains can also be represented as a matrix. For example, in the f - x domain the columns of the matrix  $\mathbf{X}$  represent each trace, but the rows represent frequency samples. Methods for noise attenuation rely on the rank reduction of the data matrix by applying a SVD. Reducing the rank of this matrices allows to identify coherent and incoherent components in the data. SVD consists basically of the decomposition of the matrix  $\mathbf{X}$  into a weighted sum of orthogonal rank one matrices, called eigenimages of  $\mathbf{X}$  (Ulrych et al., 1999). I will first introduce the SVD decomposition method, followed by its application for rank reduction of a matrix.

#### Singular Value Decomposition (SVD)

The SVD arises from the problem in linear algebra of finding the eigenvalues and eigenvectors of a matrix. Assuming a matrix  $\mathbf{C}$  that is Hermitian of size  $m \times m$ , and a vector  $\mathbf{x}$  which elements are not all zero, then the eigenvalues of  $\mathbf{C}$  are the ones that satisfy :

$$\mathbf{C}\mathbf{x} = \lambda \mathbf{x} , \qquad (2.1)$$

and the vectors  $\mathbf{x}$  are the eigenvectors of  $\mathbf{C}$ . The number of non-zero eigenvalues of  $\mathbf{C}$  represents its rank. Now, expanding this operation to matrix decomposition, the Hermitian matrix  $\mathbf{C}$  can be represented by an arrangement of it eigenvalues and eigenvectors as:

$$\mathbf{C} = \mathbf{U} \mathbf{\Lambda} \mathbf{U}^H , \qquad (2.2)$$

where the columns of **U** are the eigenvectors of **C**,  $\Lambda$  is a diagonal matrix containing the eigenvalues of **C** organized in descending order, and  $[]^H$  denotes the Hermitian or conjugate transpose of the matrix (Manning et al., 2009). Now, given that the previous decomposition is restrained for squared matrices, a different approach has to be made to decompose a rectangular matrix.

Let **X** be an  $m \times n$  matrix, with  $m \ge n$  and rank  $r \le n$ . The rank of a matrix is the number of linearly independent rows or columns, therefore, the maximum rank of a matrix is equivalent to  $rank(\mathbf{X}) \le min\{m,n\}$  (Manning et al., 2009). The application of SVD consists of decomposing this matrix as:

$$\mathbf{X} = \mathbf{U} \mathbf{\Sigma} \mathbf{V}^H , \qquad (2.3)$$

where **U** is the matrix whose columns are the eigenvectors of  $\mathbf{X}\mathbf{X}^{H}$ , **V** is the matrix whose columns are the eigenvectors of  $\mathbf{X}^{H}\mathbf{X}$  and  $\boldsymbol{\Sigma}$  is a diagonal matrix containing the singular values of **X**. The singular values of **X** are obtained from the eigenvalues of  $\mathbf{X}\mathbf{X}^{H}$  as  $\boldsymbol{\Sigma} = \sqrt{\Lambda}$ . We can relate equations 2.2 and 2.3 by assuming  $\mathbf{C} = \mathbf{X}\mathbf{X}^{H}$ , which leads to:

$$\mathbf{C} = \mathbf{X}\mathbf{X}^{H} = \mathbf{U}\mathbf{\Sigma}\mathbf{V}^{H} \ \mathbf{V}\mathbf{\Sigma}\mathbf{U}^{H} = \mathbf{U}\mathbf{\Sigma}^{2}\mathbf{U}^{H} = \mathbf{U}\mathbf{\Lambda}\mathbf{U}^{H} .$$
(2.4)

The same operation can be applied by assuming  $\mathbf{C} = \mathbf{X}^H \mathbf{X}$ . This leads to  $\mathbf{X}^H \mathbf{X} = \mathbf{V} \mathbf{\Sigma}^2 \mathbf{V}^H$ . With these operations it is clear how the singular values and singular vectors of  $\mathbf{X}$  are related to the eigenvalues and eigenvectors of  $\mathbf{X}\mathbf{X}^H$  and  $\mathbf{X}^H \mathbf{X}$ .

#### **Rank reduction**

The main characteristic of a low-rank matrix is that its elements are not independent from each other. Because of this, the problem of approximating one matrix by another, with lower rank, cannot be formulated in a straightforward manner, as a least-squares problem (Eckart and Young, 1936). Instead of a least-square inversion, one can use SVD to calculate the low rank approximation of a matrix. Let **X** be a  $m \times n$  matrix, subject to  $m \ge n$  and with rank  $r \le n$ . Let k be a real number such as k < r. The low rank approximation problem consist of finding a  $m \times n$  matrix  $\mathbf{X}_k$ , whose rank is at most k, which minimizes the Frobenius norm of the difference  $\mathbf{X} - \mathbf{X}_k$ (Manning et al., 2009). This is equivalent to:

$$\|\mathbf{X} - \mathbf{X}_k\|_F = \sqrt{\sum_{i=1}^m \sum_{j=1}^n |\mathbf{x}_{ij} - \widetilde{\mathbf{x}}_{ij}|^2} .$$
(2.5)

Eckart and Young (1936) found that this problem has a unique solution and that it can be solved by using SVD. The following steps lead to the solution of the rank approximation problem:

- 1. Decompose the initial matrix **X** by using equation 2.3, meaning  $\mathbf{X} = \mathbf{U} \mathbf{\Sigma} \mathbf{V}^{H}$ .
- 2. Replace by zero all but the first k elements of the diagonal matrix  $\Sigma$  to obtain the matrix  $\Sigma_k$ .
- 3. The resulting rank reduced matrix is obtain by  $\mathbf{X}_k = \mathbf{U} \mathbf{\Sigma}_k \mathbf{V}^H$

This process is equivalent to replacing by zero all but the first k columns of **U** and **V** and all but the first k elements of  $\Sigma_k$  and then to apply  $\mathbf{X}_k = \mathbf{U}_k \Sigma_k \mathbf{V}_k^H$ . The computation of the rank reduced matrix  $\mathbf{X}_k$  can also be calculated in a more efficient way by using the principal eigenvectors of  $\mathbf{X}\mathbf{X}^H$ , maintaining  $m \ge n$ , as (Freire and Ulrych, 1988):

$$\mathbf{X}_k = \mathbf{U}_k \mathbf{U}_k^H \mathbf{X} \ . \tag{2.6}$$

The latter allow to define the operator  $\mathbf{U}_k \mathbf{U}_k^H$  to apply the rank reduction process. The recovered matrix  $\mathbf{X}_k$  is at most rank k, and it leads to the lowest possible Frobenius norm of  $\mathbf{X} - \mathbf{X}_k$ .

We have seen how the process of rank reduction can be completed by the use of SVD. With this information it is possible to understand the principles that lay behind the rank reduction techniques for noise attenuation. These concepts are fundamental in the application of SSA.

#### 2.3 Singular Spectrum Analysis in 1-D time series

Let  $s(t) = (s_1, s_2, ..., s_N)$  be a time dependent signal, where N is the number of samples of the data. This signal is the product of a series of dynamic processes that controls the measured quantity plus noise. The application of SSA to the time series s(t) is performed as follows:

#### Embedding

SSA consists of the decomposition of the time series in its singular spectrum. This decomposition is applied to multidimensional series. It is possible to go from a one-dimensional space to a multidimensional space by using the process of embedding. This consists of decomposing the time series in a sequence of lagged vectors, which arises from the method of delays (Broomhead and King, 1986). Now, let L be the length for these lagged vectors having 1 < L < N, which is also called the embedding dimension (Elsner and Tsonis, 1996). The number of lagged vectors will depend on the embedding dimension as K = N - L + 1. Each lagged vector will have the form:

$$l_i = (s_i, s_{i+1}, \dots, s_{i+L-1})^T \quad 1 \le i \le K ,$$
(2.7)

where  $[]^T$  denotes the transpose of a matrix. The matrix that is built from the organization of the lagged vectors as  $\mathbf{M} = (l_1, l_2, ..., l_K)$  is called the trajectory matrix. The resulting trajectory matrix  $\mathbf{M}$  is:

$$\mathbf{M} = \begin{pmatrix} s_1 & s_2 & \cdots & s_K \\ s_2 & s_3 & \cdots & s_{K+1} \\ \vdots & \vdots & \ddots & \vdots \\ s_L & s_{L+1} & \cdots & s_N \end{pmatrix} .$$
(2.8)

The main characteristics of this matrix is  $\mathbf{M}_{ij} = s_{i+j+1}$ , where  $1 \leq i \leq K$  and  $1 \leq j \leq L$ . This means that the anti-diagonals of the matrix present the same values, and are symmetrical around the main diagonal. The behavior of this trajectory matrix is that of a Hankel matrix. The process of embedding can be summarized as  $\mathbf{M} = Ms(t)$ , where M is the Hankelization operator. The embedding dimension L is the main parameter to select during the embedding step. Elsner and Tsonis (1996) suggests that the results from the application of SSA are not significantly sensitive to the value of L as long as N is considerably larger than L, recommending the use of L = N/4. The selection of small values of L has the advantage of increasing the confidence in the results when the objective of the analysis present high frequencies. Other authors have said that L has to be sufficiently large so that the main behavior of the time series to analyze is content in each lagged vector (Golyandina et al., 2001). These statements show that selecting the embedding parameter involves a tradeoff between the amount of information in each vector and the confidence of the results. In the end, it is clear that this parameter can be adjusted depending on the objective of the study.

#### Singular Value Decomposition

This step consists in the decomposition of the trajectory matrix by using SVD. As explained in the previous section, SVD is a decomposition of the form:

$$\mathbf{M} = \sum_{i=1}^{r} \sqrt{\lambda_i} u_i v_i , \qquad (2.9)$$

where  $\lambda_i$  is the *i*th eigenvalue of  $\mathbf{M}\mathbf{M}^H$ , r is the rank of  $\mathbf{M}$  and  $u_i$  and  $v_i$  are the *i*th eigenvectors of  $\mathbf{M}\mathbf{M}^H$  and  $\mathbf{M}^H\mathbf{M}$ . In general,  $\sigma_i = \sqrt{\lambda_i}$  is called the singular value of the matrix  $\mathbf{M}$ . Expression 2.9 can be converted to matrix notation as:

$$\mathbf{M} = \mathbf{U} \mathbf{\Sigma} \mathbf{V}^H , \qquad (2.10)$$

where  $\Sigma$  is the diagonal matrix containing all the singular values in descending order and **U** and **V** are the matrices containing the set of orthonormal vectors  $u_i$  and  $v_i$  respectively. Given that the eigenvectors of **M** arise from the autocorrelation matrix  $\mathbf{MM}^H$ , the components that present the most coherency in the data will be weighted by singular values with higher values. This way, the decomposition of the trajectory matrix in its singular spectrum is very useful to identify trends in the data. Also, given that the signal in the time series is correlated between time lagged windows, it will be represented by the largest singular values. Because of this, singular values with less weight can be identified as noise, making possible the use of this tool in denoising the time series. It is useful to present the singular spectrum of the data as a graphical representation of the singular values of the matrix **M**. To easily visualize the contribution of each singular value, it is convenient to graph the percentage of each value compared to the sum of all the singular values.

#### Rank Reduction

Whatever the objective of the application of SSA is, the rank reduction of the trajectory matrix has to be applied. The rank reduction process was explained in the previous section. When analyzing the dynamical components of the time series, different singular values can be grouped to recover physical behaviors identified in the decomposition. For noise reduction, the rank that represents most of the signal has to be identified before the rank reduction step. In general, the process consists in recovering a small subset of singular values compared to the full rank of the trajectory matrix. Let k be the desired rank for the trajectory matrix, this can be obtained by doing:

$$\mathbf{M}_k = \mathbf{U}_k \boldsymbol{\Sigma}_k \mathbf{V}_k^T , \qquad (2.11)$$

where  $\mathbf{M}_k$  is the recovered rank-reduced trajectory matrix. The recovered matrix is rank = k and presents the lowest possible Frobenius norm.

#### **Diagonal averaging**

If the recovered matrix  $\mathbf{M}_k$  is a Hankel form, then the recovery of the time series can be done by just selecting the values in the anti-diagonals of  $\mathbf{M}_k$ . In other words, being  $s_k$  the desired recovered time series after the rank reduction of the trajectory matrix, the element n of this time series will be recovered by all the elements  $\mathbf{M}_k(i,j)$  along the secondary diagonal, being (i,j) such that i + j - 1 = n.

Regretfully, this situation rarely happens in practice. In case that the Hankel form is not preserved in the rank-reduced result, the process of recovering the time signal  $s_k$  is by averaging in the anti-diagonals of  $\mathbf{M}_k$ . Golyandina et al. (2001) introduces an operator that is helpful to describe the diagonal averaging of the recovered matrix. To simplify the explanation let's assume that  $L \leq K$ . The case where  $K \leq L$  is similar, but applying the operator to  $\mathbf{M}_k^T$ . Now, the operator works as follows: let i + j - 1 = n and N = L + K - 1, then the element n of  $s_k$  is

$$s_{k}(n) = \begin{cases} \frac{1}{n} \sum_{l=1}^{n} \mathbf{M}_{k}(l, n-l-1) & \text{for } 1 \leq n \leq L \\ \frac{1}{L} \sum_{l=1}^{L} \mathbf{M}_{k}(l, n-l-1) & \text{for } L+1 \leq n \leq K \\ \frac{1}{K+L-n} \sum_{l=n-K+1}^{L} \mathbf{M}_{k}(l, n-l-1) & \text{for } K+1 \leq n \leq N \end{cases}$$
(2.12)

The latter can be summarized as  $s_k = A\mathbf{M}_k$ , where A is the averaging over the anti-diagonals operator described by equation 2.12. This operation retrieves the component of the initial time series s that was recovered after the rank reduction of the trajectory matrix.

We have seen the main four steps to compute SSA on time series signals. The interpretation of the reconstructed components using different singular values is a topic that has been object of extensive research. For further information in the use of SSA in time series the books from Elsner and Tsonis (1996) and Golyandina et al. (2001) are recommended, which gives more details on the use of this technique. However, some examples will be presented at the end of this chapter, where the use of SSA for decomposition and noise attenuation are tested.

#### 2.4 Examples

To test the SSA algorithm for time series analysis I will present first the decomposition of a simple cosine function  $s(t) = \cos(2\pi\omega t + \phi)$ , being the temporal frequency  $\omega = 0.1$  rad/s and the phase  $\phi = 0.2$  rad. This function was contaminated with random noise with a variance of 0.5 and zero mean. Given that a cosine function can be represented by the sum of two exponentials as  $\cos(\theta) = \frac{1}{2}(e^{+i\theta} + e^{-i\theta})$ , this function is expected to have two representations with high correlation in the singular spectrum. The reason for this is explained in chapter 3.2. SSA was applied by following the four steps previously presented. The embedding dimension used to form the trajectory matrix was L = N/4, mostly because we know that there are enough cycles in this time window to perform a successful analysis. This test was repeated for L = N/3 and L = N/2, with very similar results.



Figure 2.1: Singular Spectrum for a) cosine function with no noise and b) cosine function in presence of noise.

The step of decomposing the signal using SVD is applied to the initial cosine function and to the one contaminated with noise. Figure 2.1a) shows the singular spectrum for the initial case with no noise and figure 2.1b) shows the singular spectrum for the noisy one. Notice that the initial cosine function is represented by two singular values, confirming the assumption previously made. In the presence of noise, the number of singular values different from zero increases. It is possible to observe that the first two singular values are the ones with higher energy. In general, it is possible to differentiate between the singular values that represent the signal by looking for an abrupt change on the contribution of each of them. Even if we do not have a priori information about these data, it is possible to conclude that the first two singular values represent the main oscillatory components of the signal, and the rest of them represent the noise.

Figure 2.2 shows the resulting time series after the recovery using each of the first 10 singular



Figure 2.2: Decomposition of a noisy cosine function in its singular-spetrum.

values separately. The first two columns present the original signal and the noisy one, to which SSA was applied. The amplitudes of each curve are normalized, but its contribution to the curve is proportional to the amplitude of its associate singular value, showed in the bar diagram. The graphic at the bottom shows the Fourier amplitude spectrum of each curve. In this exercise one observes the behaviour of the individual data components. The interpretation of the singular spectrum is not always easy. Sometimes there is no abrupt change in the amplitude of the singular values. In this case the selection of the final rank of the matrix will be subjected to the objectives of the study. Figure 2.3a) shows the original cosine function before the addition of noise and figure 2.3b) shows the same cosine function in the presence of random noise. Figure 2.3c) is the result of the application of SSA to the noisy data, performing the rank reduction with the first two singular values. It can be

observed that the process was successful in removing the noise, also attenuating slightly the amplitudes of the result, compared to the original one. This example shows how SSA can be a powerful tool for noise removal in time series analysis.



Figure 2.3: Result from filtering the noisy cosine function using SSA. a) Cosine function with no noise, represents the expected solution. b) Cosine function contaminated with random noise. c) Result of filtering using SSA. The decrease in amplitude in the solution is due to the large amount of noise in the data.

SSA is now applied to the Wolf sunspot number curve from 1700 to 1998. In this case, there is no previous knowledge of the trends or behavior of this time series. The goal is not to use SSA to denoise the curve, but to analyze the components of the data. Anyways, some comparison can be done with the previous example. In this test, the embedding dimension is set to L = N/2. Given that the amount of data for this series is higher; the amount of time lagged windows will be larger, which will maintain the confidence in the result. The selection of this embedding dimension makes the trajectory matrix square. In time series analysis it is common to subtract the mean of the curve to each value of the record. This has been the objective of standardizing the observations when the singular spectrum from different records are compared. Elsner and Tsonis (1996) proved that after subtracting the mean to the elements of the curve, the singular values of the time series remain unchanged. For the analysis of the Wolf sunspot number curve, the mean was subtracted from the record to avoid including a very low frequency spectrum. These low frequencies tend to overwhelm the higher frequencies of interest in the Fourier domain.

Figure 2.4 shows the singular spectrum for the sunspot number series. One observes that the decrease in the contribution of each singular value is smooth. Unlike the singular spectrum



Figure 2.4: Singular Spectrum for the Wolf sunspot number time series.

for the noisy cosine function, it is hard to differentiate the main components of signal from the ones of the noise. Despite this, it is possible to use SSA to identify components with high coherency in the data. Figure 2.5 shows the results of the decomposition of the Wolf number series, presenting the contribution of the first 15 singular values. The first column is the original record after subtracting the mean. The Fourier amplitude spectrum is also shown in the bottom of the figure for each of the components. The amplitudes for each curve are normalized, and the contribution of each component is proportional to the weight of the singular value, which is shown in the top bar graph. We can see that the original record presents a dominant frequency between 0.08 and 0.1 cycles/year. This frequency content is recovered by the first four singular values, showing that those are the components with higher energy in the data. The periodicity shown by the first two components correspond to, approximately, 11 years/cycle, which is known as the solar cycle (Wilson, 1994). The next singular values show components with lower and higher frequency contents. The analysis and forecasting of the Wolf numbers series using SSA has been presented by Loskutov et al. (2001), who expands on the advantages and disadvantages of SSA for the analysis of solar activity data. The interpretation of the physical processes that influence these components will not be discussed given that they are out of the scope of this thesis. From this example we can extract that the decomposition of a time series can provide information about the processes that influence time records. We can also see how the singular spectrum of some time series is smooth, in which case the selection of the final rank to filter the data is not a simple task.



Figure 2.5: Decomposition of the sunspots number curve in its singular-spectrum.

## CHAPTER 3

# Singular Spectrum Analysis for noise attenuation in seismic records

#### 3.1 Backgound

This chapter presents the use of SSA for random noise filtering in pre-stack and post-stack seismic data. Although the application of SSA in time series analysis has been studied for a long time, its use in seismic data processing is rather recent. Trickett (2002) introduced the use of a rank reduction method called f - x eigenimage filter, which is based in the work of Cadzow (1988). Trickett (2008) suggested that the method should be called Cadzow filtering, in honor to the author from whom they based the technique. The application of the Cadzow method is documented in Trickett and Burroughs (2009). The Cadzow method and SSA are equivalent, but they arise from different fields of study. We have seen that SSA was developed for the analysis of time series, while the Cadzow method was proposed as a technique for the denoising of images (Cadzow, 1988). The relationship between the Cadzow method and SSA is presented by Sacchi (2009), who denominated the technique as f - x SSA.

Filtering random noise in seismic records involve the application of SSA in the f - x domain (Trickett, 2008; Sacchi, 2009). In the first part of this chapter, SSA is applied to one single frequency at the time, assuming it as a vector that varies in space. The methodology applied here is analogous to the time series examples shown in chapter 2. The results from the noise attenuation achieved by SSA are compared with those from using f - x deconvolution. The latter is the standard technique for random noise attenuation. Although the results show

no evidence of a significant improvement on the amount of noise attenuated by SSA over the f - x deconvolution, it does a better task in preserving the signal.

A significant improvement on the application of SSA for random noise attenuation arises from its expansion to multiple dimensions, which is called MSSA. This chapter describes the process of expanding SSA for the analysis of a 3-D seismic data set, which involves the application of MSSA in two dimensions (2-D MSSA). MSSA is also applied in the f - x domain. The main difference between SSA and 2-D MSSA is that the two spatial dimensions of a 3-D seismic record are used simultaneously for the analysis. An example of the application of 2-D MSSA is presented. The latter shows that this expansion improves significantly the attenuation of random noise. The results from 2-D MSSA are compared with those from applying f - x deconvolution and SSA. The expansion to MSSA is then generalized to Ndimensions (N-D MSSA). Although the theory behind N-D MSSA is explained here, it was not tested with synthetic or real examples. The design of an example applying MSSA to more than two dimensions is beyond the scope of this thesis, and it is strongly recommended for future research.

### 3.2 Singular Spectrum Analysis in seismic data processing

The application of SSA in seismic data processing is similar to the one applied for the analysis of time series. The main difference is that SSA is applied in the f - x domain of the seismic records. Instead of using a temporal vector as input for the analysis, it uses a spatial vector. Therefore, two extra steps have to be added to the SSA application sequence described in chapter 2. These steps consist of converting the input data from the t - x domain to the f - x domain and back. The application of SSA for the attenuation of random noise in seismic records is performed as follows:

#### Application of a Fourier transform to each channel:

We start our discussion by considering a 2-D waveform with constant dip. The latter is analogous to a single event in a seismic section. For simplicity, one can imagine a portion of a seismic waveform seen in a small window of analysis. This waveform can be represented as:

$$s(x,t) = w(t - px)$$
, (3.1)
where x denotes space, t time, p dip and w(t) is a pulse or wavelet. Figure 3.1 shows a graphic representation of this 2-D waveform.



Figure 3.1: Example of a 2-D waveform with constant dip.

This signal can be converted to the f - x domain by applying a Fourier transform to each channel of the 2-D waveform. The data in the f - x domain are represented by the following expression:

$$S(x,\omega) = A(\omega)e^{-i\omega px} , \qquad (3.2)$$

where  $\omega$  denotes temporal frequency. Let's consider in addition that the spatial variable x is replaced by its discrete counterpart  $x = n\Delta x$ , with n representing the channel number. Also, without losing generality,  $S(x_n, \omega) = S_n$ . It is clear that the following analysis is valid for one monochromatic temporal frequency  $\omega$ . It is easy to demonstrate that adjacent channels (at a given frequency) must obey a linear recursion. Let us first rewrite equation 3.2 using the previous assumptions:

$$S_n = A e^{-i\omega p n \Delta x} . \tag{3.3}$$

Similarly, it is possible to use the same notation for the previous channel n-1 as:

$$S_{n-1} = Ae^{-i\omega p(n-1)\Delta x} = Ae^{-i\omega pn\Delta x}e^{-i\omega p\Delta x}.$$
(3.4)

Substituting equation 3.3 in 3.4 we obtain that channel  $S_n$  is related to the previous channel  $S_{n-1}$  as:

$$S_n = PS_{n-1} aga{3.5}$$

where  $P = e^{i\omega p\Delta x}$ .

It is possible to demonstrate that multiple 2-D waveforms also obey a linear recursion at a given frequency. For example, a record presenting two 2-D waveforms with different dips can be represented in the f - x domain as:

$$S_n = A_1 e^{-i\xi_1 n} + A_2 e^{-i\xi_2 n} = S_n^{(1)} + S_n^{(2)} , \qquad (3.6)$$

where  $A_k$  are the amplitudes for each event k, and  $\xi_k = \omega p_k \Delta x$ , being  $p_k$  the dip of each event. It is clear that if the dips of each event are different, then  $\xi_1 \neq \xi_2$ . In a similar way as in equation 3.4, one can represent the two previous channels n-1 and n-2 as:

$$S_{n-1} = A_1 e^{-i\xi_1(n-1)} + A_2 e^{-i\xi_2(n-1)}$$
  

$$S_{n-2} = A_1 e^{-i\xi_1(n-2)} + A_2 e^{-i\xi_2(n-2)} .$$
(3.7)

Using equation 3.5 one can form the following system of equations:

a)  
b)  
c)  

$$\begin{cases}
S_n = S_n^{(1)} + S_n^{(2)} \\
S_{n-1} = a_1 S_n^{(1)} + a_2 S_n^{(2)} \\
S_{n-2} = a_1^2 S_n^{(1)} + a_2^2 S_n^{(2)}
\end{cases}$$
(3.8)

where  $a_k = e^{-i\omega p_k \Delta x}$ . The solution to this system of equations can be obtained by organizing equations 3.8a) and 3.8b) in their matrix form:

$$\begin{bmatrix} S_{n-1} \\ S_{n-2} \end{bmatrix} = \begin{bmatrix} a_1 & a_2 \\ a_1^2 & a_2^2 \end{bmatrix} \begin{bmatrix} S_n^{(1)} \\ S_n^{(2)} \end{bmatrix} .$$
(3.9)

Since  $\xi_1 \neq \xi_2$  the matrix  $\begin{bmatrix} a_1 & a_2 \\ a_1^2 & a_2^2 \end{bmatrix}$  is invertible, so the solution for this system is  $\begin{aligned} S_n^{(1)} &= \alpha S_{n-1} + \beta S_{n-2} \\ S_n^{(2)} &= \gamma S_{n-1} + \nu S_{n-2} \end{aligned}$ (3.10)

Finally, substituting equation 3.10 in equation 3.8a) we obtain the linear relationship between  $S_n$ ,  $S_{n-1}$  and  $S_{n-2}$ , which is

$$S_n = P_1 S_{n-1} + P_2 S_{n-2} . aga{3.11}$$

This relationship shows the linear recursion between adjacent channels in the presence of two events. It is possible to expand this relation for k events. It is also important to mention that this recursion is the basis for f - x deconvolution and represents the predictability of the signal in the f - x domain (Sacchi and Kuehl, 2001; Ulrych and Sacchi, 2005). This predictability is the key element in the success of SSA for random noise attenuation.

### Embedding of each frequency into a Hankel matrix:

Let  $\mathbf{S}_{\omega} = [S_1, S_2, S_3, ..., S_{N_x}]^T$  be a spatial vector of a given frequency  $\omega$  from the f - x domain. Here  $N_x$  represents the number of space samples of the data. This spatial vector  $\mathbf{S}_{\omega}$  is analogous to the time series analyzed in chapter 2, but in this case it components are complex numbers. We apply SSA using four steps similar to the ones described in chapter 2. The spatial vector  $\mathbf{S}_{\omega}$  is embedded into a Hankel matrix of the form:

$$\mathbf{M} = \begin{pmatrix} S_1 & S_2 & \cdots & S_{K_x} \\ S_2 & S_3 & \cdots & S_{K_x+1} \\ \vdots & \vdots & \ddots & \vdots \\ S_{L_x} & S_{L_x+1} & \cdots & S_{N_x} \end{pmatrix},$$
(3.12)

where the length of the lagged vectors  $L_x$  is the parameter that controls the matrix dimensions. Building a square Hankel matrix is a common strategy when SSA is applied to seismic records (Trickett, 2008). The latter can be achieved by setting the lagged vectors length as  $L_x = floor(N_x/2) + 1$ . By doing this, the number of columns in the Hankel matrix would be  $K_x = N_x - L_x + 1$ . Expression 3.5 imposes a linear relationship between the columns of the Hankel matrix **M** as:

$$\mathbf{M} = \begin{pmatrix} S_1 & PS_1 & \cdots & P^{K_x - 1}S_1 \\ S_2 & PS_2 & \cdots & P^{K_x - 1}S_2 \\ \vdots & \vdots & \ddots & \vdots \\ S_{L_x} & PS_{L_x} & \cdots & P^{K_x - 1}S_{L_x} \end{pmatrix}.$$
 (3.13)

It is easy to observe that for a simple f - x signal, the Hankel matrix reduces to a matrix with rank = 1. It is clear that in the presence of uncorrelated noise the rank of the matrix will increase. If the record contains two events, equation 3.11 shows that all the columns of the matrix **M** are linear combinations of the first two columns

$$\mathbf{M} = \begin{pmatrix} S_1 & S_2 & P_1 S_2 + P_2 S_1 & \cdots & P_1 S_{n-1} + P_2 S_{n-2} \\ S_2 & S_3 & P_1 S_3 + P_2 S_2 & \cdots & P_1 S_{n-1} + P_2 S_{n-2} \\ \vdots & \vdots & \vdots & \ddots & \vdots \\ S_{L_x} & S_{L_x+1} & P_1 S_{L_x+1} + P_2 S_{L_x} & \cdots & P_1 S_{n-1} + P_2 S_{n-2} \end{pmatrix} .$$
(3.14)

For the superposition of k events with constant dip, one can show that the Hankel matrix is rank = k. This means that, by knowing the number of events contained in the initial data set, one can know the minimum rank of the matrix that represents all the events. As a consequence, the selection of the final rank of the matrix is not subjective, representing an advantage over the application of SSA in time series analysis, and over many other rank reduction methods for noise attenuation in seismic records. Rank reduction via the Singular Value Decomposition (SVD) of **M** can be used to capture the singular-vectors that model the signal.

### Decomposition of the Hankel matrix using Singular Value Decomposition (SVD):

The singular value decomposition of **M** is given by:

$$\mathbf{M} = \mathbf{U} \mathbf{\Sigma} \mathbf{V}^H , \qquad (3.15)$$

where:

 $\mathbf{U} = \text{eigenvectors of } \mathbf{M}\mathbf{M}^H$ 

 $\mathbf{V} = \text{eigenvectors of } \mathbf{M}^H \mathbf{M}$ 

 $\Sigma$  = singular values of **M** in descending order.

This process has been developed in chapters 2, so no further explanation is needed.

### Rank Reduction of the Hankel matrix:

The noise in the data  $(\mathbf{S}_{\omega})$  can be removed by using a low-rank reconstruction of the matrix **M**. As seen in chapter 2, the rank reduction of the Hankel matrix can be obtained by recovering a subset of its singular values as:

$$\mathbf{M}_k = \mathbf{U}_k \boldsymbol{\Sigma}_k \mathbf{V}_k^H , \qquad (3.16)$$

where  $\Sigma_k$  indicates the diagonal matrix containing the first k largest singular values of **M**.

### Averaging on the anti-diagonals of the recovered Hankel matrix:

To recover the filtered data we average along the anti-diagonals of the matrix  $\mathbf{M}$  (Sacchi, 2009). This process is achieved by using equation 2.12, from chapter 2. Equation 3.17 provides a visual example of the averaging along the anti-diagonals of a Hankel matrix built from a vector with seven entries.

$$\mathbf{M}_{k} = \begin{pmatrix} S_{1} & S_{2} & S_{3} & S_{4} \\ & \swarrow & & \swarrow \\ S_{2} & S_{3} & S_{4} & S_{5} \\ & \swarrow & & \swarrow \\ S_{3} & S_{4} & S_{5} & S_{6} \\ & \swarrow & & \swarrow \\ S_{4} & S_{5} & S_{6} & S_{7} \end{pmatrix} .$$
(3.17)

Using equation 2.12 is equivalent to  $\hat{\mathbf{S}}_{\omega} = A\mathbf{M}_k$ , where  $\hat{\mathbf{S}}_{\omega}$  is the filtered version of  $\mathbf{S}_{\omega}$ . The averaging step recovers a filtered version of a single frequency of the f - x domain. One can easily implement SSA denoising by applying the rank reduction technique for each individual frequency components  $\omega$ .

#### Application of an inverse Fourier transform to each channel:

After the rank reduction process is applied to each individual frequency  $\omega$ , the resulting data in the f - x domain is taken to the t - x domain by applying an inverse Fourier transform to each channel. With the application of this step, the filtered image is recovered.

These six steps summarize the application of SSA for random noise attenuation in seismic records. Its application is simple compared to other methods and it presents the advantage of allowing us to select the final rank of the Hankel matrix objectively. This algorithm can be extended to work in multiple domains, which will be studied in the next section.

# 3.3 2-Dimensional Multichannel Singular Spectrum Analysis (2-D MSSA)

Previously I discussed the application of SSA to attenuate random noise in a two dimensional seismic record. In this section I will study the expansion of SSA to multiple dimensions. This expansion is called MSSA. It was presented by Read (1993) in the context of time series analysis and by Trickett (2008) for noise attenuation in seismic records. In this section, MSSA technique is studied for random noise attenuation of a 3-D seismic record. Given

that SSA works on each frequency at the time, this would involve the application of a two dimensional MSSA (2-D MSSA). The application of MSSA follows the same six steps presented previously for SSA, but the construction of the Hankel matrix is expanded in order to add all the dimensions. For this reason, this section is not divided into the MSSA steps. Instead, it focuses on describing the construction of this new Hankel matrix.



Figure 3.2: Example of one frequency slice organized as a matrix from a 3-D record to perform 2-D MSSA.

Let's consider a 3-D waveform with constant dip, where the z axis represents time and x and y axis represents space dimensions. These data are transformed to the f - x domain by applying a Fourier transform to each channel of the cube. This way, the 3D data depends on x, y and a temporal frequency  $\omega$ . For one frequency slice the data can be organized in a matrix as follows:

$$\mathbf{S}_{\omega} = \begin{pmatrix} S(1,1) & S(1,2) & \cdots & S(1,N_y) \\ S(2,1) & S(2,2) & \cdots & S(2,N_y) \\ \vdots & \vdots & \ddots & \vdots \\ S(N_x,1) & S(N_x,2) & \cdots & S(N_x,N_y) \end{pmatrix} .$$
(3.18)

The number of traces in the x and y dimensions are given by  $N_x$  and  $N_y$ , respectively. The extraction of  $\mathbf{S}_{\omega}$  from the 3-D seismic record in the f - x domain is shown schematically in figure 3.2. 2-D MSSA first construct one Hankel matrix for each inline (x) component of  $\mathbf{S}_{\omega}$ . In other words,

$$\mathbf{M}_{j} = \begin{pmatrix} S(1,j) & S(2,j) & \cdots & S(K_{x},j) \\ S(2,j) & S(3,j) & \cdots & S(K_{x}+1,j) \\ \vdots & \vdots & \ddots & \vdots \\ S(L_{x},j) & S(L_{x}+1,j) & \cdots & S(N_{x},j) \end{pmatrix} .$$
(3.19)

Again, a good strategy is to build a square Hankel matrix, by setting the length of the lagged vector in x as  $L_x = floor(N_x/2) + 1$  and  $K_x = N_x - L_x + 1$ . Next, to add the cross-line (y) dimension to the analysis, we construct a Hankel of Hankel matrix, which consists of the inline Hankel matrices organized on a block Hankel matrix as:

$$\mathbf{M} = \begin{pmatrix} \mathbf{M}_1 & \mathbf{M}_2 & \cdots & \mathbf{M}_{K_y} \\ \mathbf{M}_2 & \mathbf{M}_3 & \cdots & \mathbf{M}_{K_{y+1}} \\ \vdots & \vdots & \ddots & \vdots \\ \mathbf{M}_{L_y} & \mathbf{M}_{L_{y+1}} & \cdots & \mathbf{M}_{N_y} \end{pmatrix}.$$
 (3.20)

Equation 3.20 is equivalent to equation 3.12 but now expanded to the y dimension. This size of the block Hankel matrix  $\mathbf{M}$  is  $(L_y \times L_x) \times (K_y \times K_x)$ . Figure 3.3 presents an example of the construction of a block Hankel matrix. In here we can appreciate how the size of the resulting Hankel matrix is much larger than the size of the input matrix. It is evident that the size of the block Hankel matrix depends on the number of channels in each dimension of the data.



Figure 3.3: Construction of a Block Hankel matrix.

In the previous section we observed that there exists a linear relation between the columns of each Hankel matrix  $\mathbf{M}_{j}$ . It is possible to expand this analysis for a 3-D waveform in the frequency domain and, this way, find a relationship between the columns of the block Hankel matrix. A 3-D waveform with constant dip can be represented in time as:

$$s(x, y, t) = w(t - p_x x - p_y y) , \qquad (3.21)$$

which preserves the nomenclature of equation 3.1 and includes y as the second dimension in space. The dip of the event in the x and y dimensions are represented by  $p_x$  and  $p_y$ . This waveform is represented in the f - x domain by the following expression:

$$S(x, y, \omega) = A(\omega)e^{-i\omega(p_x x + p_y y)} .$$
(3.22)

In a similar way as in equation 3.3, it is possible to replace the spatial variables x and y by its discrete counterpart  $x = n\Delta x$  and  $y = m\Delta y$ , with n and m representing the channel number in x and y respectively. Also, without losing generality,  $S(x_n, y_m, \omega) = S_{nm}$ . Applying the analysis from equations 3.3 and 3.4 it is possible to find a linear relationship between adjacent channels in dimension y as:

$$S_{n(m-1)} = Ae^{-i\omega p_x n\Delta x} e^{-i\omega p_y (m-1)\Delta y} = Ae^{-i\omega p_x n\Delta x} e^{-i\omega p_y m\Delta y} e^{-i\omega p_y \Delta y} .$$
(3.23)

It is clear from equation 3.23 that there is a linear relationship between channel m and the previous channel m-1 in the dimension y, for all the elements n in the dimension x as:

$$S_{nm} = QS_{n(m-1)} , (3.24)$$

where  $Q = e^{i\omega p_y \Delta y}$ . Equation 3.24 presents the relationship between each component of the Hankel matrix  $\mathbf{M}_j$  with those from the previous Hankel matrix  $\mathbf{M}_{j-1}$ , which finally produce a linear relation between the columns of the block Hankel matrix (equation 3.20) as:

$$\mathbf{M} = \begin{pmatrix} \mathbf{M}_{1} & Q\mathbf{M}_{1} & \cdots & Q^{K_{y}-1}\mathbf{M}_{1} \\ \mathbf{M}_{2} & Q\mathbf{M}_{2} & \cdots & Q^{K_{y}-1}\mathbf{M}_{2} \\ \vdots & \vdots & \ddots & \vdots \\ \mathbf{M}_{L_{y}} & Q\mathbf{M}_{L_{y}} & \cdots & Q^{K_{y}-1}\mathbf{M}_{L_{y}} \end{pmatrix}.$$
(3.25)

The rank of a block Hankel matrix has been studied in detail by (Hua, 1992) and (Yang and Hua, 1996). From here on, the procedure ends with the last four steps for the application of SSA on seismic records, presented in the previous section. The block Hankel matrix is decomposed via SVD, using equation 3.15. Then, the rank of the Hankel matrix is reduced using equation 3.16. Next, the filtered data are retrieved by properly averaging along the anti-diagonals of each individual Hankel matrix composing the low-rank approximation of

the block Hankel matrix. The latter is important since, if the average is calculated for all the anti-diagonals of the block Hankel matrix, different entries would be mixed in the operation, resulting on a poor recovery of the solution. Finally, the 2-D MSSA technique is applied to all the frequencies of the data and an inverse Fourier transform is calculated to convert the solution from the f - x domain to the t - x domain.

The results from the application of the 2-D MSSA are significantly better than those of SSA, as it is shown in section 3.5. This is a consequence of the addition of more information to the analysis. The improvement in noise attenuation makes 2-D MSSA a very useful tool in seismic data processing. The main problem found in the application of 2-D MSSA is the large amount of computational time required for it to work. This happens while applying the SVD decomposition of the block Hankel matrix for the rank reduction step. SVD is a very expensive algorithm when it is applied to large matrices. The latter could be a problem in 2-D MSSA given that the rank reduction process has to be repeated for all the frequencies. Although 2-D MSSA can be significantly slower than other noise attenuation techniques, the improvement in the results obtained for random noise filtering justifies further research on this technique.

# 3.4 N-Dimensional Multichannel Singular Spectrum Analysis (N-D MSSA)

Previously, the application of SSA and 2-D MSSA was presented. This two applications work in dimensions that are easy to imagine, and that can be easily identified in a data set. But in seismic data processing it is possible to analyze more than 3 dimensions of information. To clarify this, let us review the different dimensions that can be found in a seismic data set. A 1-Dimension record would be a single trace in time. The application of SSA to this type of record is carried on in the time domain, and was analyzed in chapter 2. A 2-Dimension data set can be a shot gather, which presents the information of several traces versus one spatial dimension. It varies with time and records the information generated by one source. Another example of a 2-D record is a post-processing stacked gather, which also present several traces in one spatial dimension, depending on time. The random noise attenuation of 2-D seismic data is performed by applying SSA in the f - x domain. The seismic data in 3-Dimensions comes, for example, from gathering several shots, producing a cube which dimensions are shot number, offset and time. In other words, it has two spatial dimensions plus time. Another example of a 3-D seismic data is a post-processing stacked data, which dimensions are x and y axis, plus time. The attenuation of random noise in 3-D seismic data can be achieved via 2-D MSSA in the f - x domain, as described in the previous section.

Although the addition of more than three simultaneous dimensions in seismic data analysis is not necessarily intuitive, it is possible. These dimensions are the result of the geometry of a 3-D seismic survey. A 3-D seismic data set is commonly represented as a 5-Dimensional volume, which requires two spatial dimensions to identify the source location, two more to identify the receiver location and a fifth dimension, which is time. Another way of representing the data in 5-D is by identifying the two dimensions of the source-receiver midpoint location, the source-receiver distance (offset) and the azimuth of arrival as the third and fourth dimension, and finally the time, as the fifth dimension (Trad, 2009). The information contained by each of this dimensions can be utilized simultaneously in the N-D MSSA analysis. The process is exactly the same as for 2-D MSSA, with the only difference that with the addition of each new dimension, a larger block Hankel matrix is constructed. The process can be summarized as follows:

- 1. Apply a Fourier transform to all traces of the N-Dimensions to convert the data to the frequency domain.
- 2. For a single frequency, the process of embedding is applied by building a block Hankel matrix using the N-Dimensions. This means building Hankel of Hankel of Hankel matrix N times for the N-Dimensions. This step is an expansion to N-D of equation 3.20 and figure 3.3. Trickett and Burroughs (2009) show that when SSA is applied for N-Dimensions, the block Hankel matrix keeps presenting rank = k when there is a presence of k independent linear events in the data.
- 3. The block Hankel matrix is decomposed using SVD from equation 3.15.
- 4. The rank of the block Hankel matrix is reduced by applying equation 3.16.
- 5. The N-Dimensions are recovered by averaging in the anti-diagonals of the individual Hankel matrices that construct the block Hankel matrix. Similarly to 2-D MSSA, if all the items in the anti-diagonals of the block Hankel matrix are averaged, it would result in a poor solution.
- 6. This process is repeated for all the frequencies of the data, followed by the application of an inverse Fourier transform to convert the data from the frequency domain to the time domain.

The addition of more information may improve significantly the results of N-D MSSA for random noise attenuation. Also, its implementation is simple compared to the expansion of other techniques to multiple dimensions. Despite of this, its application can be expensive in computational time. The construction of examples to test the use of N-D MSSA is beyond the objectives of this thesis, so no results are displayed here. The information presented in this section may be used as the initial step for further work in this topic.

# 3.5 Results and Discussion

Several tests are presented in this section in order to evaluate the SSA algorithms. First, SSA is tested on synthetic gathers, which present 3 linear events with different dips, and 3 hyperbolic events with different curvatures. Then, it is applied to a real stacked gather contaminated with random noise. In these examples, the results from SSA are compared to those obtained from applying an f - x deconvolution filter. Finally, the 2-D MSSA method is applied to a 3-D synthetic gather which present 4 events with different dips. This example is compared with the results from using f - x deconvolution and 1-D SSA as random noise attenuation techniques in each slice of the seismic cube.

### 3.5.1 SSA

### Synthetic data

If the seismic record to analyze contains k different linear events, it is only necessary to recover k singular values to represent the data when SSA is applied. An example of this is showed in figure 3.4, where f - x deconvolution and SSA are applied to a noiseless data set, presenting 3 events with different dips and amplitudes. In this example, figure 3.4a) is the result of the f - x deconvolution filter, figure 3.4b) is the result from the application of SSA, figure 3.4c) is the original data prior to noise contamination and figures 3.4d) and 3.4e) are the noise estimators resulting from subtracting the filtered data from the noiseless data. The noise estimators in figures 3.4d) and 3.4e) present mostly zero values in its traces, meaning that neither f - x deconvolution nor SSA distort the events when it is applied to pure signal. In this example, only the first 3 singular values where recovered in the application of SSA. The contribution of each singular value for each frequency of the data in showed in figure 3.5. We can see that there are only 3 singular values larger than zero.

The next example shows the same data used in the previous test, but contaminated with random noise (figure 3.6). In the presence of noise, the number of non-zero singular values increases. This difference can be observed by comparing the number of singular values that represent the noiseless data (figure 3.5) with the ones that represent the data plus noise (figure 3.7). It is clear that the rank of the Hankel matrix increased with the addition of noise. In this example, f - x deconvolution and SSA are applied to attenuate this random noise. The results of this example are shown in figure 3.6, where figure 3.6a) shows the result from the application of f - x deconvolution, figure 3.6b) shows the result from the use of SSA and figure 3.6c) presents the noisy input data. Figures 3.6d) and 3.6e) are the



Figure 3.4: Application of SSA on a noiseless record. a) f-x deconvolution filtering. b) SSA filtering using one frequency at a time. c) Original data prior to noise contamination. d), e) Difference between the filtered data and the noise-free data.



Figure 3.5: Singular Values for the noiseless data.

noise estimators resulting from subtracting the initial data (3.6c)) to each of the filter results 3.6a) and 3.6b) respectively. Figure 3.6f) is the original data prior to noise contamination and figures 3.6g)and 3.6h) are the difference between the noiseless data 3.6f) and the results from each filter 3.6a) and 3.6b).



Figure 3.6: Application of SSA on a record contaminated with random noise. a) f - x deconvolution filtering. b) SSA filtering using one frequency at a time. c) Noisy input data. d), e) Noise estimators for a) and b). f) Original data prior to noise contamination. g), h) Difference between the filtered data and the noise-free data.

Noise reduction via f - x deconvolution is achieved by setting the number of frequencies used in the analysis to  $N_f = 8$ , the pre-whitening parameter is  $\mu = 0.1$ , and the initial and final frequencies for the analysis are  $f_{init} = 1$  Hz and  $f_{final} = 80$  Hz respectively. For the application of SSA, the final rank of the Hankel matrix is k = 3 and the initial and final frequencies are  $f_{init} = 1$  Hz and  $f_{final} = 80$  Hz, respectively. The selection of the f - xdeconvolution parameters is achieved by keeping the amount of filtered signal minimum in the noise estimators. The only parameter to select for SSA is the final rank of the matrix. In this case this parameter is k = 3, giving that the data present 3 events with different dips.



Figure 3.7: Singular Values for the data contaminated by random noise.

We can observe in figure 3.6 that SSA yields to very similar results than f - x deconvolution. It is important to notice, however, that SSA preserves the entire signal, while f - xdeconvolution seems to attenuate slightly the amplitude of the seismic events.

An example of the application of f - x deconvolution and SSA for random noise attenuation in a record presenting hyperbolic events is also presented. Given that f - x deconvolution and SSA work under the assumption of linear events, they must be applied using windows in space when the events are curved. In short windows it is possible to consider a curved event as linear. Figure 3.8 shows the results for this test. As in figure 3.6, figure 3.8a) shows the result from the f - x deconvolution filter using spatial windows and figure 3.8b) shows the result from the 1-D SSA filter using windows. Next, figure 3.8c) shows the noisy input data, followed by figures 3.8d) and 3.8e) which present the noise estimators resulting from the subtraction of the initial data in figure 3.8c) and the filter results from figures (3.8a) and (3.8b) respectively. Finally, figure (3.8f) represents the original data prior to noise contamination and figures 3.8g) and 3.8h) are the difference between the noiseless data (figure 3.8f)) and the results from the application of the filters, figures 3.8a) and 3.8b), respectively. The initial data had 120 traces and 1 second in time. The windows are selected to cover the entire data in time, with 1 second height and 14 traces width, overlapping every 4 traces. We observe in figure 3.8 that both filters succeed in attenuating some random noise from the data. The amount of noise filtered by applying SSA (figure 3.8b)) appears to be similar to the amount of noise attenuated by applying f - x deconvolution (figure 3.8a)). It is important to notice that the difference between the noiseless data and the result of SSA (figure 3.8h)) reflect a complete preservation in the amplitudes of the signal, while the application of f - x deconvolution presents the attenuation of part of the signal. The result of this example shows that, although SSA does not present a significant advantage



Figure 3.8: Results using windows in space on hyperbolic events: a) f - x deconvolution filtering. b) SSA filtering using one frequency at the time. c) Noisy input data. d), e) Noise estimators for a) and b). f) Original data prior to noise contamination. g), h) Difference between the filtered data and the noise-free data.

on the amount of random noise removed compared to f - x deconvolution, it displays the advantage of maintaining intact the amplitudes of the events.

### Field data

One of the main steps during seismic data processing is the stacking of traces of a common mid-point (CMP) to obtain a seismic section. Even if the process of stacking is, by itself, a very powerful filter, it still presents important components of random noise. In figure 3.9 we can observe the result of the application of f - x deconvolution and SSA filters for random

noise attenuation of an stacked gather, constructed from real data. In here, figure 3.9 a) shows the initial noisy gather, while figure 3.9 b) presents the result after f-x deconvolution and figure 3.9 c) shows the result after applying SSA. If the events of the stacked data are linear, SSA and f - x deconvolution can be applied in the whole record. However, if the events present some curvature, or the stacked record is large, it is recommended to apply the filters using windows in space. This is the case for this example, where the size of the record demanded the use of spatial intervals of analysis.

Both methods are applied on overlaying windows of data. The latter aims to denoise sections that present similar characteristics. The record presents 500 traces and 2.2 seconds in total. The window size is 8 traces, overlapping 10 traces, for a total of 28 traces in space by 0.260 seconds with an overlapping of 0.08 seconds, for a total of 0.420 seconds in time. The parameters for the application of f - x deconvolution are selected in order to minimize the amount of signal attenuated by it. The best filtering using f - x deconvolution for this example requires a number of frequencies of  $N_f = 4$ , a pre-whitening parameter of  $\mu = 0.001$ and an initial and final frequencies of  $f_{init} = 1Hz$  and  $f_{final} = 100Hz$  respectively. All the events in each overlapping window have very similar dips, so they are recovered by the same singular value. Because of this, the final rank of the Hankel matrices in the application of SSA was set to k = 1. The analysis of SSA was carried on between the frequencies  $f_{init} = 1Hz$  and  $f_{final} = 80Hz$ .

The results of this test show that SSA attenuates a similar amount of noise than f - x deconvolution. This is consistent with the results from the linear synthetic results. To better appreciate the behavior of these filters, figure 3.10 shows a zoom in the black box of figure 3.9. We can observe that both filters attenuate part of the noise, being the results very similar. We can also see that the noise estimator does not present coherent events attenuated by the filters.

### 3.5.2 2-D MSSA

### Synthetic data

The next example shows the expansion of SSA to 2-D MSSA, which involves the analysis of two spatial dimensions simultaneously. The input data is a synthetic 3-D seismic cube, presenting 4 linear events in the x and y dimensions and which is contaminated with random noise. The objective of this test is to compare the advantages in noise attenuation of 2-D MSSA over one dimensional noise attenuation methods, such as f - x deconvolution and SSA.

The results from the application of the noise attenuation techniques to the input record are presented in figure 3.11. We can observe the initial data contaminated with noise (figure



Figure 3.9: Results of the application of f - x deconvolution and SSA to a stacked section with random noise. a) Initial data. b) Random noise attenuation using f - x deconvolution. c) Noise attenuation via SSA.



Figure 3.10: Data in the Black box of Figure 3.9.





Figure 3.11: Noise attenuation using 2-D MSSA. a) Input data. b) Noiseless data representing the expected solution. c) Noise attenuation applying f - x deconvolution. d) Noise attenuation applying SSA. e) Noise attenuation applying 2-D MSSA.

3.11a)) and an image of the same input data with no noise, which is the expected result of the filtering (figure 3.11b)). Figure 3.11c) presents the result of applying f - x deconvolution for random noise attenuation on the 3-D data. To achieve better results, the technique is first

applied to each slice of the x dimension, then it is applied to each slice in the y dimension and finally the two results are averaged. This way, the technique takes advantage of the predictability of the signal in both dimensions. Random noise attenuation using SSA follows the same procedure and it is applied in both dimensions independently. The results obtain by applying SSA are presented in figure 3.11d). It is clear that the final signal-to-noise ratio achieved by using f - x deconvolution and SSA are very similar, which supports the results from the tests on 2-D seismic data. Figure 3.11e) shows the result obtained from the application of 2-D MSSA for random noise attenuation to the 3-D seismic cube. As we know, this method uses the information of both dimensions simultaneously by constructing a block Hankel matrix in the SSA analysis. It is clear that 2-D MSSA removed significantly more noise than f - x deconvolution or SSA. Furthermore, the signal to noise ratio has improved considerably and the events remain intact if this result is compared to the expected answer in figure 3.11b).



Figure 3.12: Slice in the y = 50 m of figure 3.11. a) Input data. b) Noise attenuation by using f - x deconvolution. c) Noise attenuation using SSA. d) Noise attenuation using 2-D MSSA. e) Noiseless data representing the expected solution. f), g) and h) are the result of the subtraction of filter results (b), c) and d)) from the noiseless data (e)), respectively.

To perform a better analysis of the results of this test, figures 3.12 and 3.13 show a slice on y = 50 m and x = 30 m respectively. The location of these slices is shown on each cube of figure 3.11, identified with a solid line. Figures 3.12 and 3.13 present the same



Figure 3.13: Slice in the x = 30 m of figure 3.11. a) Input data. b) Noise attenuation by using f - x deconvolution. c) Noise attenuation using SSA. d) Noise attenuation using 2-D MSSA. e) Noiseless data representing the expected solution. f), g) and h) are the result of the subtraction of filter results (b), c) and d)) from the noiseless data (e)), respectively.

results from two different perspectives, but their characteristics are similar and can be interpreted together. The slice a) in both figures shows the noisy input data and the slice e) presents the noiseless data. The slices b), c) and d) show the results from the application of f - x deconvolution, SSA and 2-D MSSA respectively. Finally, slices f), g) and h) are the difference of the results shown in b), c) and d) with the noiseless data from e). This difference allows identifying the amount of noise remaining in the results, as well as if some signal was attenuated. As seen before, the amount of noise attenuated by the f - xdeconvolution and SSA are very similar, but we can see that the f - x deconvolution distorts the amplitudes of the signal. Nevertheless, the result from applying 2-D MSSA shows a much better attenuation of the random noise, leaving the signal intact. These results evidence the benefits of including several dimensions simultaneously in the 2-D MSSA technique.

## 3.6 Summary

In this chapter, SSA was expanded for the attenuation of random noise in seismic records. For this, it was necessary to apply SSA to each frequency, requiring two extra steps in its application to convert the data. The six steps to apply SSA in the f - x domain were described, together with an explanation of the selection of the final rank of the Hankel matrices. SSA was tested in two synthetic records contaminated with random noise and presenting linear and hyperbolic events. Its results were compared to the one from applying f - x deconvolution, which is a traditional technique for random noise attenuation. The amount of noise attenuated by SSA showed to be similar to results obtained from applying f - x deconvolution. Nevertheless, f - x deconvolution appeared to distort slightly the amplitudes of the events, while SSA maintained the signal untouched.

The expansion of SSA to multiple dimensions was also covered, which was called MSSA. The application of this expansion was explained by adding two dimensions simultaneously (2-D MSSA) and then generalized to N dimensions (N-D MSSA). An example was presented testing the application of 2-D MSSA for random noise attenuation. This example compared the results of 2-D MSSA with the ones of applying f - x deconvolution and SSA. Noise attenuation using 2-D MSSA showed to improve significantly the signal-to-noise ratio of a 3-D seismic record contaminated with noise, outperforming f - x deconvolution or SSA. The expansion to N-D MSSA was not tested giving that it is beyond the scope of this thesis. For the rest of this thesis MSSA will refer to 2-D MSSA.

# CHAPTER 4

# Fast application of Multichannel Singular Spectrum Analysis by randomization

# 4.1 Motivation and Background

The application of MSSA yield to very promising results for random noise attenuation. Its simplicity when extended to several dimensions also gives a large advantage over traditional noise attenuation techniques. What makes the method less attractive is the amount of computations and running time necessary for its application. It is very slow when applied to large matrices. This increase on time happens in the rank reduction step (during the application of the SVD). The running time of the SVD algorithm increases very fast for large matrices. When MSSA is applied, the number of elements in the Hankel matrix of one frequency increases rapidly with the addition of channels in each dimension. Giving that SSA is applied to one frequency at the time, the running time will be multiplied by the amount of frequencies in the analysis, adding substantially to the total cost of the method. This limits the amount of information that can be included in the analysis and motivates the research of faster and accurate algorithms for rank reduction.

Rank reduction techniques are used in multiple disciplines, giving that large matrices with low rank appear in many scientific fields. Aiming to solve some of these problems, many authors have proposed different mathematical methods which present fewer amounts of operation and preserves accuracy over the traditional SVD algorithm. These algorithms are general, since they are designed to work in many different problems, no matter the scientific area. One of these techniques is the pivoted QR factorization, which consists in the decomposition of the initial matrix  $\mathbf{M}$  in a orthonormal matrix  $\mathbf{Q}$  and an upper triangular matrix  $\mathbf{R}$  (Gu and Eisenstat, 1996). This decomposition was introduced by Golub (1965) and some of its algorithms have been compiled by Golub and Van Loan (1996). This technique is fast compared to the traditional SVD algorithm, but its accuracy depends on how quickly the singular spectrum decays (Rokhlin et al., 2009). However, we have seen that in the presence of noise the singular spectrum has a smooth decay, which can make the QR factorization not suitable for MSSA. Another technique for rank reduction is the Lanczos Method, which involves a partial tridiagonalization of the initial matrix (Golub and Van Loan, 1996). This method has been applied by Trickett (2003) to accelerate the rank reduction of an eigenimage noise attenuation technique. Although the Lanczos algorithm is effective in reducing the computational time of the SVD, it can be delicate to implement and, like the QR factorization, its accuracy is subjected to how smooth the singular spectrum of the matrix is (Halko et al., 2009).

A relatively new approach to the rank reduction problem is the application of randomized algorithms. The goal of these techniques is to use a subset of the initial matrix to calculate the SVD on a smaller number of elements than if the whole matrix was used (Drinea et al., 2001). In general, all randomized algorithms follow the same basic steps (Halko et al., 2009):

- 1. Apply a preprocessing to the initial matrix.
- 2. Use random techniques to take random samples from the data.
- 3. Apply a post-processing to the random samples to obtain the final low rank approximation.

It is believed that the initial steps on randomized algorithms where presented by Papadimitriou et al. (1998), with its application to latent semantic indexing (LSI) (Halko et al., 2009). Works on structured dimension reduction of matrices, presented by Martinsson et al. (2006), Sarlos (2006) and Rudelson and Vershynin (2007), among others, set the foundations for low-rank approximation algorithms. These algorithms have been developed by Woolfe et al. (2008), Liberty et al. (2007) and Rokhlin et al. (2009). A recent review and expansion of these techniques have been presented by Halko et al. (2009), where they explain in simple terms the process of rank reduction via randomization.

In this chapter we propose the use of an algorithm based on the work of Rokhlin et al. (2009). Overall, this process entails computing the SVD of randomly compressed data matrices. The advantage of this algorithm arises from adding a power iteration to the dimension reduction step, which improves its performance. In essence, the algorithm replaces performing the SVD of large matrices by the application of the SVD on two reduced matrices. The latter leads to an algorithm that is well suited for denosing problems where one makes extensive use of the SVD. This is often used when noise attenuation is applied to a large number of overlapping spatio-temporal windows. Bear in mind that the acceleration strategies for computing the SVD that are proposed here are applicable to any rank reduction filtering method. I show here how this novel technique on rank reduction improves the efficiency of MSSA by preserving the quality of the results.

## 4.2 Five step algorithm for Random SVD

The randomized algorithm studied by Rokhlin et al. (2009) is summarized as follows. Let  $\mathbf{M}$  be a complex  $m \times n$  block Hankel matrix, where  $m \leq n$ . Let  $\widetilde{\mathbf{M}}_k$  be the rank k desired approximation to the original matrix  $\mathbf{M}$ . In addition, l is an oversampling integer where l > k and  $l \leq m - k$ . Rokhlin et al. (2009) recommend  $l \geq 2k$ . It can be shown that the first k singular values and singular vectors of  $\mathbf{M}$  can be approximated from the SVD of the smaller matrix  $\mathbf{P} = \mathbf{R} [\mathbf{M} \mathbf{M}^H]^i \mathbf{M}$ , where  $\mathbf{R}_{(l \times m)}$  is a matrix of independent and identically distributed (i.i.d) Gaussian numbers with zero mean and unitary variance. The process requires the following algorithm (Liberty et al., 2007; Rokhlin et al., 2009)

- 1. Compute  $\mathbf{P}_{(l \times n)} = \mathbf{R}_{(l \times m)} [\mathbf{M}_{(m \times n)} \mathbf{M}_{(n \times m)}^H]^i \mathbf{M}_{(m \times n)}$ .
- 2. Use the SVD of the reduced matrix  $\mathbf{P}_{(n \times l)}^{H}$  to obtain

 $\mathbf{P}_{(n \times l)}^{H} = \mathbf{Q}_{(n \times n)} \boldsymbol{\rho}_{(n \times l)} \mathbf{L}_{(l \times l)}^{H}.$ 

3. Use the first k columns of  ${\bf Q}$  to compute

$$\mathbf{S}_{(m \times k)} = \mathbf{M}_{(m \times n)} \mathbf{Q}_{(n \times k)} \,.$$

4. Form the SVD of the matrix **S** 

$$\mathbf{S}_{(m \times k)} = \mathbf{U}_{(m \times m)} \mathbf{\Sigma}_{(m \times k)} \mathbf{T}_{(k \times k)}$$

5. Compute the product

$$\mathbf{V}_{(n imes k)} = \mathbf{Q}_{(n imes k)} \mathbf{T}_{(k imes k)}$$

Then,

$$\widetilde{\mathbf{M}}_{k\ (m\times n)} = \mathbf{U}_{(m\times m)} \mathbf{\Sigma}_{(m\times k)} \mathbf{V}_{(k\times n)}^{H} .$$

$$(4.1)$$

The rank-k matrix  $\mathbf{M}_k$  satisfy the condition,

$$\left\| \mathbf{M} - \widetilde{\mathbf{M}}_k \right\|_2 \le C m^{1/(4i+2)} \sigma_{k+1} , \qquad (4.2)$$

where  $\sigma_{k+1}$  is the k + 1st singular value of **M** and *i* is a power iteration of the algorithm. This expression shows the accuracy of the method. Rokhlin et al. (2009) states that *C* is a constant independent of **M** that depends on the parameters of the algorithm, and is at the very least C < 10. The error of the algorithm is reduced for smaller matrices and larger values of the power iteration variable *i*.

The algorithm requires  $\mathcal{O}(nmki)$  floating point operations, which makes it faster than the traditional SVD algorithm, which requires  $\mathcal{O}(nm^2)$  floating point operations (Rokhlin et al., 2009). For large Block Hankel matrices, *i* has to be large to maintain accuracy. Given that the number of computations will increase for higher values of *i*, the selection of this variable will require a tradeoff between accuracy and computational expense.

# 4.3 Methodology

The previous randomized algorithm for rank reduction is tested for MSSA. Eleven 3D windows of different sizes contaminated with random noise are used to evaluate the changes in running time for each algorithm. Each window consists of 3 linear events with different dips and amplitudes. The number of channels in each window increase as  $N_x = 21 + 4j = N_y$ , with j = 0, 1, 2, ..., 10. The algorithms are written in Matlab 7.3. SVD is computed using the Matlab function svd which uses the ZGESVD driver from Anderson et al. (1999). For this test, a workstation with 2GB of RAM and an AMD Athlon(tm) 64 X2 Dual Core Processor 3800+ was used.

Rank reduction in MSSA is computed using the traditional SVD algorithm and the randomized algorithm proposed by Rokhlin et al. (2009), setting the power iteration variable to i = 1, 2 and 3. The latter is to test the differences in accuracy and computational time when using different values of i. For every test the parameters are set to k = 3 and l = 6, giving that there are 3 events with different velocities and amplitudes in the synthetic. The computational time of the algorithm is measured by starting when the data is converted to the f - x domain (beginning of the MSSA process) and finishing after the result is returned to the t - x domain (output of MSSA). The measured time takes into account the rank reduction of the Block Hankel matrix for each temporal frequency. In this example the analysis is done on 328 frequencies, from 0.24 Hz to 80 Hz. Together with the computational time the signal-to-noise ratio (S/N) of the results is calculated by using the expression:

$$S/N = 10 \log_{10} \left( \frac{\|d_0\|_2^2}{\|d_f - d_0\|_2^2} \right) , \qquad (4.3)$$

where  $d_0$  is the noiseless data and  $d_f$  is the result after applying MSSA. These allow testing

how fast the algorithms are and the accuracy of the results.

## 4.4 **Results and Discussion**

Table 4.1 shows the results for the 11 windows. We can see how the number of entries in the Block Hankel matrix becomes very large when the size of the window increases. For better visualization, the results for computational time and S/N ratio are presented in figures 4.1 and 4.2.

Number of traces	Size of Block Hankel Matrix	SVD		R-SVD $i = 1$		R-SVD $i = 2$		R-SVD $i = 3$	
$N_x \times N_y$	$m \times n$	time (s)	S/N (dB)	time (s)	S/N (dB)	time (s)	S/N (dB)	time (s)	S/N (dB)
$21 \times 21$	$121 \times 121$	48.85	4.48	16.18	5.14	21.14	5.27	22.45	5.05
$25 \times 25$	$169 \times 169$	143.87	5.94	34.38	6.44	43.85	6.79	52.67	6.66
$29 \times 29$	$225 \times 225$	359.22	7.28	86.18	7.43	107.14	7.99	128.93	7.95
$33 \times 33$	$289 \times 289$	848.3	8.22	245.12	8.53	292.43	9.11	375.45	8.94
$37 \times 37$	$361 \times 361$	1777.54	9.34	623.74	9.37	613.94	10.25	805.63	10.17
$41 \times 41$	$441 \times 441$	3282.82	10.01	1192.91	10.01	1343.81	10.91	1534.48	10.85
$45 \times 45$	$529 \times 529$	5809.06	10.78	2126.43	10.73	2395.07	11.6	2684.21	11.58
$49 \times 49$	$625 \times 625$	9246.18	11.4	3418.46	11.26	3886.72	12.13	4336.99	12.13
$53 \times 53$	$729 \times 729$	14679.52	11.91	5379.91	11.69	6414.98	12.75	7224.39	12.75
$57 \times 57$	$841 \times 841$	23898.37	12.59	8425.84	12.19	9553.47	13.35	10560.69	13.38
$61 \times 61$	$961 \times 961$	34955.46	12.01	12459.66	11.49	13715.46	12.91	15003.76	12.99

Table 4.1: Computing times and S/N ratio for noise attenuation of different sizes of data windows  $(N_x \times N_y)$ . SVD means the application of multichannel Singular Spectrum Analysis (MSSA) denoising using the standard Singular Value Decomposition. R-SVD are results for the randomized SVD algorithm described in the text.

Figure 4.1 presents the computational time for the traditional SVD and the randomized algorithm with i = 1, 2 and 3. It is evident that the randomized SVD algorithm is faster than the traditional SVD algorithm. This improvement represents an approximate reduction of 50% over the traditional SVD computational time. For high values of i in the randomized SVD, the computational time increases.

Figure 4.2 presents the S/N ratio in dB of the results for MSSA using the different algorithms. The S/N ratio of the traditional SVD is assumed to be the ideal result for MSSA. The randomized algorithm using i = 1 shows a decrease in the S/N ratio compared to the traditional SVD when the number of rows m in the Block Hankel matrix is high. The other two curves, representing the random SVD with i = 2 and i = 3, maintain a S/N ratio very similar to the traditional SVD, showing that the results are consistent.

Figures 4.3 and 4.4 show the application of MSSA for the window with  $N_x = 61$ . They present a slice in x = 31 and y = 31 respectively. We can see that the randomized algorithm with i = 1 removes part of the signal, supporting the results from Figure 4.2.



Figure 4.1: Plot showing the increments of the computational time vs. the number of columns of the Hankel matrix (m).



Figure 4.2: Plot showing the Signal to Noise ratio depending on the number of columns of the Hankel matrix (m).



Figure 4.3: Slice in x = 31 for the data with size  $61 \times 61$ . a) Initial noisy data. b) Result using the traditional SVD. c), d) and e) Results using the random SVD algorithm with i = 1, i = 2 and i = 3 respectively. f) Noiseless data  $(d_0)$ . g), h) i) and j) are the subtraction of f) from b), c), d) and e) respectively.

# 4.5 Summary

The application of rank-reduction denoising is limited by the computational cost of traditional SVD algorithms. The application of a randomized SVD algorithm to improve the computational time of MSSA was presented. The results show that the randomized SVD yields an approximately 50% gain in efficiency over the traditional SVD. The accuracy of the randomized algorithm depends on the size of the block Hankel matrix (m) and on the selection of the parameter i. The use of larger values of i improves the accuracy of the randomized SVD but increases the number of calculations. Because of this, the selection of the parameter i is a tradeoff between accuracy and speed. Alternatively, one can increase the final rank of MSSA leaving i = 1. By doing this some amount of noise will be recovered, but no signal will be filtered. In this case, we can achieve good accuracy for the MSSA result maintaining a lower amount of calculations.

Overall, this algorithm has proved successful in accelerating MSSA. It is clear that the use of the randomized SVD can be applicable to any rank reduction filtering method. Impor-



Figure 4.4: Slice in y = 31 for the  $61 \times 61$  data. a) Initial noisy data. b) Result using the traditional SVD. c), d) and e) Results using the random SVD algorithm with i = 1, i = 2 and i = 3 respectively. f) Noiseless data  $(d_0)$ . g), h) i) and j) are the subtraction of f) from b), c), d) and e) respectively.

tant computing time savings are attainable when the problem requires constructing Hankel matrices from data that depends on three or four spatial dimensions. In this case we will need to form Hankel matrices of sizes that are unmanageable by standard SVD algorithms.

# CHAPTER 5

# Interpolation Using Multichannel Singular Spectrum Analysis

# 5.1 Background

Seismic data acquisition consists in generating a wave field in the surface of a survey area and then in extracting the information that is reflected from the subsurface geology. The information is recorded by receivers placed at the land surface. This means that the information coming from the reflections will be recorded at discrete points in the space. Although seismic surveys are designed to maintain a regular grid of sources and receivers, this will rarely happen due to logistic or economical constrains. Because of this, seismic data may be irregularly sampled in space or presenting gaps where no traces are recorded. Many seismic processing tools for noise attenuation or imaging require the input data to be sampled regularly in space to work properly. Different techniques have been developed to regularize data and to recover missing traces. These techniques commonly require the conversion of the data into different domains by using methods like the Fourier transform (Liu and Sacchi, 2004), Radon transform (Trad et al., 2002) or the Curvelet transform (Herrmann and Hennenfent, 2008). One of the methods that interpolate traces in the Fourier domain is the one developed by Spitz (1991) and expanded by Porsani (1999) and Naghizadeh and Sacchi (2010), which applies a prediction error filter in the f - x domain of the data. This method is very powerful when interpolating undersampled and aliased data in a regular grid; but interpolating traces in an irregular pattern can be more difficult (Abma and Kabir, 2006).

An application that aims to solve the interpolation of missing traces in an irregular pattern was developed by Abma and Kabir (2006). They apply an iterative method that consists in thresholding the frequency spectrum after a 2D Fourier transform. Subsequently, they replace the recovered traces in the original gather to perform the analysis again. After some iterations, the traces would be recovered. The amount of missing traces and the differences in amplitudes of the events control the number of iterations to recover the events with the correct amplitude. This algorithm is called Projection onto convex sets (POCS) and was developed by Youla and Webb (1982) for image restoration problems.

The objective of this chapter is to study the application of MSSA to interpolate missing traces in an irregular pattern. Just like the interpolation method developed by Spitz (1991), MSSA works in the f - x domain of the data, but instead of performing a prediction error filter, it relies in the rank reduction of Hankel matrices. The process is improved by using the POCS algorithm applied by Abma and Kabir (2006), and instead of thresholding the frequency spectrum we apply MSSA. This chapter aims mainly in understanding the processes that allow MSSA to interpolate seismic data and looks to be a guideline on how to apply it. Our results are not compared with results of other interpolators. This comparison is strongly recommended for future research in SSA.

# 5.2 Application

The process of interpolating and recovering missing traces in seismic record using MSSA is the same as filtering random noise. In fact, both processes can be carried out simultaneously. As discussed in chapter 3, the application of SSA to filter random noise in seismic records consists in six steps that are:

- 1. Fourier transform to take the data to the f x domain
- 2. Embedding of each frequency into a Hankel matrix
- 3. Decomposition in its singular spectrum via SVD
- 4. Rank reduction of the Hankel matrix
- 5. Averaging in the Hankel matrix anti-diagonals
- 6. Inverse Fourier transform to return to the time domain.

A noise free record with a single linear event will lead to a Hankel matrix with rank one. Similarly, a number of k linear events with different apparent velocities will result in a Hankel matrix with rank k (see equation 3.13). In the presence of noise the rank of the Hankel matrix will increase. The rank reduction step of SSA aims to filter random noise by recovering a Hankel matrix with the rank that represents the events contained in the record. This procedure is still valid in the presence of missing traces, as long as they are in a regular grid. The missing samples will be treated as noise by SSA, and the rank reduction step will recover some of the amplitudes expected on these missing traces. This allows SSA to be a powerful interpolator that can be easily extended to several dimensions.

Missing traces in a seismic record are characterized by presenting zeros in all its samples. When transformed to the f - x domain, these missing traces still present zero values in each frequency. These samples become part of the Hankel matrix after the embedding step. In the presence of missing samples, the rank of the Hankel matrix increases. This is analogous with the case where random noise contaminates the record. The rank reduction process will approximate the Hankel matrix to the expected low-rank matrix that represents the signal with no gaps. With this, the missing samples are replaced by the value that generates the best low-rank approximation to the Hankel matrix. Given that this process is repeated for a range of frequencies, the missing traces will recover part of its amplitude spectrum. After one iteration, linear events are interpolated through the missing traces, but they will present lower amplitudes compared to the original events.

To recover the right amplitude for the interpolated traces, it is possible to apply the POCS algorithm utilized by Abma and Kabir (2006). This iterative algorithm is applied in the f-x domain and works using SSA or MSSA. For SSA the input data is a vector and for MSSA the input data is a matrix of a single frequency. The following description of the algorithm assumes a 2-D input data **S**. Initially, an operator T is created to identify the presence of traces on the spatial position (i, j) of the data, being T(i, j) = 1 in the cells that present a trace and T(i, j) = 0 in the cell where there are missing traces. The application of the operator T to the data **S** produces the observed data  $\mathbf{S}_{obs}$ . In other words,  $T \odot \mathbf{S} = \mathbf{S}_{obs}$ . It is evident that  $T \odot \mathbf{S}_{obs} = \mathbf{S}_{obs}$ . The difference between an operator I with ones in every cell (i, j), meaning I = ones(dim T), and T result in an operator that identify the spatial position of the missing traces. This operator is used to extract the recovered traces after one iteration, and to place them in the original input data. Then MSSA is applied again. The POCS algorithm used to interpolate data using MSSA can be summarized as follows:

for 
$$p = 1$$
: Niter  
for  $f = f_{init}$ :  $f_{final}$   
 $\mathbf{S}_p(f) = \mathbf{S}_{obs}(f) + (I - T) \odot A[\mathbf{U}_k \mathbf{U}_k^T H(\mathbf{S}_{p-1}(f))]$  (5.1)  
end  
end.

where  $f_{init}$  and  $f_{final}$  are the initial and final frequencies to analyze in the MSSA process, H is the Hankelization operator,  $\mathbf{U}_k \mathbf{U}_k^T$  is the rank reduction operator and A is the averaging

in the anti-diagonals operator, all defined in chapter 2.  $\mathbf{S}_p$  is the solution after each iteration. The operator  $\odot$  represents the Hadamard product of two matrices, which is the elementwise matrix product (Kolda and Bader, 2009). After some iterations, the algorithm will converge and the amplitudes recovered for the missing traces will be consistent with the ones present in the original traces.

An important drawback of this method is the amount of calculations that takes part in the application of MSSA for each iteration. This obstacle can be sorted by including the randomized SVD (R-SVD) described in chapter 4 to speed up the rank reduction step. By doing this the running time of the algorithm is reduced considerably, which allows increasing the number of iterations or the amount of traces to be analyzed at one time. All the examples presented here are calculated using R-SVD.

## 5.3 **Results and Discussion**

### 5.3.1 Synthetic data Example

The iterative interpolation algorithm using MSSA is first tested in a 3D synthetic data set. It presents three linear events with different dips and amplitudes. The dimension of this data set is 24 by 29 traces in the x and y spatial dimensions respectively, and 0.5 seconds in time. This is a reasonable size for a window analysis of a larger data set. The initial synthetic data is shown in figure 5.1a). The operator T was then applied to the seismic cube to decimate the data, randomly extracting 58% of the total amount of traces (figure 5.1b)). This is the input to the MSSA process. A total of 6 iterations was necessary to recover the amplitudes of the missing traces. The MSSA algorithm reduced the Hankel matrix of each frequency to a matrix of rank k = 3 giving that the data present 3 events with different apparent velocities. As explained before, the rank reduction step in MSSA is applied using the R-SVD function described in chapter 4. The result of the interpolation is presented in figure 5.1c). It is clear that the missing traces are recovered completely. Figure 5.1d) shows the difference between the expected result (figure 5.1a) and the result of the interpolation (figure 5.1c)), that may show events that are not successfully interpolated. In figure 5.1d) we can observe that there are almost no differences between the interpolation results and the initial data, so we can conclude that the method recovered almost perfectly the missing traces.

Figure 5.2 presents the same results for this example, but from a slice in channel y = 10. In here figure 5.2a) shows a slice in the original data, figure 5.2b) is the data with traces 2, 5-8, 13, 15, 17-19, 21 and 24 missing, figure 5.2c) is the result of the interpolation and figure 5.2d) is the noise estimator. The amplitudes of the interpolated traces are consistent with



Figure 5.1: Interpolation of a noiseless synthetic example cube presenting 3 events. x and y are the two spatial dimensions. a) Initial data. b) Data decimated by the operator T. It presents 58% of random missing traces. c) Result of the interpolation using MSSA. d) Difference between the result (c) and the initial data (a).



(a) (b) (c) (d) Figure 5.2: Slice in y = 10 of the noiseless synthetic example for data interpolation. a) Initial data. b) Data decimated by the operator T. c) Result of the interpolation using MSSA. d) Difference between (a) and (c).

the amplitudes of the initial traces, and the correlation of the event is maintained. These results show that the interpolation using MSSA yields to good results in recovering missing traces in linear events with different dips.

The second example shows the same synthetic data as before, but contaminated with random noise (figure 5.3a)). This test allows examining the interpolation using MSSA, as well as the noise attenuation capabilities of the method. Like in the previous example, the operator T is applied to decimate the data, extracting the same 58% of the total amount of traces (figure

5.3b)), which becomes the input to the algorithm. This example also required 6 iterations to recover the traces, by reducing the rank of each Hankel matrix to k = 3. The result of the interpolation is shown in figure 5.3c), and its difference with the original data is presented in figure 5.3d). We can see how the interpolation algorithm using MSSA successfully recovers the missing traces. In addition, it also attenuates the random noise present in the original data. In figure 5.3d) we can see how all the difference is related to the random noise. This means that all the events where recovered and filtered satisfactorily.



Figure 5.3: Interpolation of a synthetic example cube presenting 3 events and contaminated with random noise. x and y are the two spatial dimensions. a) Initial data. b) Data decimated by the operator T. It presents 58% of random missing traces. c) Result of the interpolation using MSSA. d) Difference between the result (c) and the initial data (a).

Like in the first example, the results from the interpolation on a noisy record are showed on a slice in channel y = 10 (figure 5.4). This image supports the results obtained previously, displaying a good interpolation of the events while attenuating the random noise significantly.

### 5.3.2 Real Data Example

The following example tests the MSSA interpolation algorithm in 15 common depth points (CDP) gathers, with variable number of channels. A CDP gather presents the traces that are reflected in the same point in space. This means that each trace comes from different combinations of sources and receivers and, in this case, it is equivalent to the common midpoint (CMP). The distance between source and receiver that generates each trace can vary, as long as the mid-point between them is the CDP. This means that the offset of each trace does not always follow a regular pattern. This increases with logistic constrains,



(a) (b) (c) (d) Figure 5.4: Slice in y = 10 of the synthetic example contaminated with random noise for data interpolation. a) initial data. b) Data decimated by the operator T. c) Result of the interpolation using MSSA. d) Difference between (a) and (c).

during the seismic survey, that requires a change in the location of sources or receivers. Some applications, like pre-stack migration, requires the traces to be regularly spaced in all CDP's (Naghizadeh and Sacchi, 2010). This example examines the interpolation/reconstruction of the traces missing in a group of CDP gathers after organizing their traces in a regular grid. To improve the lateral correlation of the events the CDP gather are corrected by normal move out (NMO), which horizontalize the hyperbolic events.



Figure 5.5: Initial distribution of offsets in each CDP.

Plotting each CDP versus the offset of each of its traces reveals an irregular distribution (figure 5.5). The process of regularizing these traces start with the selection of a desired grid in which we want to organize them. In this example the traces are arranged in a grid


Figure 5.6: Offsets regularized on a desired grid. Cells showing a star contain a trace, while the empty ones present missing traces.

with a cell size of 50 meters, starting in 0 meters of offset and finishing in 2350 meters. It is evident from figure 5.5 that, after dividing the area in a grid, some present several traces while some other will remain empty. When a cell contains several traces, these are averaged to obtain only one trace. The result of organizing the traces in a regular grid is shown in figure 5.6, where the cells with a star contain live traces and the empty ones contain traces which samples are all zeros. The ratio of missing traces is approximately 51%. The resulting regular data are the input to the interpolation algorithm. In this case, the iterative MSSA algorithm is designed to reduce the rank of the Hankel matrix of each frequency to k = 2. This example requires 3 iterations to converge to a solution in which the amplitudes of the missing traces are preserved.



Figure 5.7: Computational times for the use of SVD and R-SVD in the rank-reduction step of the MSSA algorithm

Given that this is a real data set, the CDP gathers present random noise that that should be attenuated. All this characteristics in the data makes possible to apply all the techniques studied in the previous chapters. The MSSA algorithm is capable to attenuate the random noise in a multi-dimensional data set as well as to interpolate missing traces after several iterations. This test compares also the use of SVD and R-SVD algorithms in the rank reduction step to analyze the validity in the use of the randomized algorithm for the interpolation process. Figure 5.8 shows the results after the iterative interpolation algorithm is applied to the CDP gathers organized in a regular grid. Figure 5.8a) shows the initial noisy cube of data, which presents missing traces. Figure 5.8b) shows the result of the interpolation using the SVD algorithm in the rank reduction step and figure 5.8c) shows the result after using the R-SVD algorithm. Both results are very similar, but R-SVD works in approximately 40% of the SVD time (Figure 5.7). These results show that the algorithm is successful in interpolating the missing traces, giving continuity to the strongest reflectors. The amplitudes of these recovered traces are consistent with those of the initial traces. The results show a dipping event on the farther offsets, which amplitudes are lower than the main events. It is possible that this event represents ground roll that is also interpolated by the algorithm. Figure 5.9 shows a slice on CDP = 11 for the initial data (a), the SVD result (b) and the R-SVD result (c). Here the recovery of the missing traces is more obvious. We can see that the interpolated events present good continuity.



Figure 5.8: Interpolation of a real cube of 15 CDP gathers regularized on a desired grid. a) Initial data after regularization. b) Interpolated data using the iterative MSSA algorithm applying the traditional SVD algorithm for the rank reduction step. c) Interpolated data using the iterative MSSA algorithm with the R-SVD technique.



Figure 5.9: Slice on CDP= 11 of the real cube interpolation example. a) initial data after regularization. b) Interpolated data using the iterative MSSA algorithm applying the traditional SVD algorithm for the rank reduction step. c) Interpolated data using the iterative MSSA algorithm with the R-SVD technique.

#### 5.4 Summary

Several seismic processing techniques require the data to have a regular spatial distribution to work properly. This condition is not always met due to economical and logistical constrains. In dimensions where the data are regularly organized many traces are missing. In other dimensions where traces are not regularly distributed the binning (gridding) step can leave spaces with missing traces. The regularization and interpolation of this traces is an important topic in seismic data processing and many algorithms have been proposed to solve this problem. The use of MSSA as a technique to interpolate traces was studied in this chapter. The algorithm proposed here is iterative and is based in the POCS algorithm used by Abma and Kabir (2006). Interpolating via MSSA was tested in a 3D synthetic data set presenting linear events. It was also tested in a 3D real data set sorted in the CDP-offset dimension.

The field data example also shows that MSSA is capable of attenuating random noise during the interpolation process. Finally, the results show that using the R-SVD algorithm in the rank reduction step improves the computational time of the interpolation and denoising process without affecting the final results. In the examples showed in this chapter we applied the 2-D MSSA algorithm, but future research efforts will focus on the design of an N-D interpolator.

### CHAPTER 6

# Singular Spectrum Analysis applied to ground roll attenuation

#### 6.1 Introduction

Previous chapters expand on how rank reduction methods are successful in attenuating random noise. These methods take advantage on the lateral uncorrelated nature of random noise. On the other hand, coherent noise correlates laterally, meaning that different denoising techniques have to be applied to filter them. Coherent noise arises from secondary wave fields generated by the source or from the ground response. Among coherent noises we can identify linear noise, reverberations and multiples. One of the linear coherent noises is the ground roll (GR), which is the vertical component of Rayleigh waves (Karsli and Bayrak, 2004). In general, surface waves are those which travel through a free surface, being confined to a layer which thickness is comparable to its wavelength (Rayleigh, 1885). In seismic data processing there is a great interest in attenuating the GR. This arises from the effect that the GR has in contaminating the seismic reflectors, decreasing significantly the quality of the data.

GR is clearly visible in seismic records as a dispersive linear noise with high amplitudes, low frequencies and low phase and group velocities. These high amplitudes and low frequencies are the reasons why the GR masks the reflections. Research has been made in the use of source and receiver patterns during data acquisition. The use of geophone patterns is a common practice to attenuate GR in land seismic surveys (McKay, 1954). This method takes advantage of the propagation direction of the GR. It consists in placing equally spaced

receivers, aligned with the source, in a distance related to the wave length of the GR. The reflections are expected to arrive vertically to each array, registering the same arrival time for each of the receivers. Since the GR travels along the surface, each geophone will record different amplitudes. The information from each receiver is then averaged. Giving that the total length of the array is a multiple of the wave length of the GR, the latter is attenuated after the averaging of the data recorded by each geophone. Unfortunately, this method only filters specific wave numbers, depending on the length of the geophone array. Early studies also expand on the use of source patterns to cancel the amplitudes of the ground roll, either by using pattern of shot holes (McKay, 1954) or by designing different sweeps and cross-correlation methods when using Vibroseis (Coruh and Constain, 1983). Application of GR attenuation techniques during data acquisition is limited. It may decrease the resolution of the signal when the source and receiver arrays overlap (Coruh and Constain, 1983) or attenuate shallow reflectors that do not arrive vertically to the receiver arrays (Knapp, 1986). This makes necessary the application of different techniques during the data processing.

Classical methods exploit the spatial and frequency differences between the GR and the reflections. Filtering the data in the f - k domain is of common use to attenuate the ground roll (March and Bailey, 1983; Duncan and Beresford, 1994). This method consists in applying a 2-D Fourier transformation in time and space to map linear events in the f - kdomain. The low frequency and low wavenumber characteristics of the GR can be identified in this domain. Thus a rejecting window can be designed to filter the GR (Yilmaz, 2001). The use of an f - k filter has the drawback of distorting the signal wavelets and introducing artifacts to the record. Another common way to filter GR, mostly during fast track data processing, is by the use of a low pass filter. The method starts with the application of the Fourier transform to each trace of the data to obtain its amplitude spectrum. A filter window is then applied to the low frequencies that represent the GR. The low frequency noise is recovered by applying the inverse Fourier transform. The recovered low frequency gather containing the GR is then subtracted from the original data. This method is successful when the main frequencies of the ground roll are very different from those of the signal. When the frequencies of the ground roll and the reflections overlap, the use of a low pass filter also eliminates low frequencies from the reflections. In this case the method becomes less effective.

This chapter expands on the use of SSA to improve the results from the application of a low pass filter during the GR attenuation process. The low frequency data recovered from the application of a low pass filter is decomposed into its singular spectrum by using SSA in the f - x domain. The same methodology used in chapter 3 is applied to build a Hankel matrix for each frequency. We then recover the signal using the subset of singular values that model only the GR. By doing this, the final low frequency gather that is subtracted from the original data will contain only the GR, solving the problem of eliminating low frequencies

in the reflections. A similar method has been applied with the name of Spectral Matrix Decomposition. This technique differentiate events by decomposing the covariance matrix of a single frequency of the f - x domain, while SSA does it on a Hankel matrix of each frequency. Mari and Glangeaud (1990) used Spectral Matrix Decomposition to differentiate arrivals in Vertical Seismic Profiles (VSP).

#### 6.2 Theory

Previous chapters showed that linear events in a seismic section could be recovered by the retrieval of a subset of its singular values after SSA. When dealing with random noise, this operation takes advantage of the incoherency of the noise between traces. In general, the first singular values will represent the signal and the amount of noise removed will depend on how many singular values are recovered. The minimum singular values to ensure that all the linear events are recovered are the same as the number of events with different dips present in the record. In the case of coherent noise attenuation the situation is more complex. In this case, the event that we wish to retain and the event that is considered noise are recovered by the first few singular values. The problem is to know if different singular values recover different events. If they do so, it is also important to find out which singular value represents each event. Analyzing how the events are projected over the orthonormal basis generated by the singular vectors holds the answer to this query.

In chapter 3 we showed that SSA is applied in the frequency domain of the data, which is shown in equation 3.2 for a single event. When more than one event is present, the data in the frequency domain can be written as

$$\mathbf{S}(x) = \sum_{i=1}^{k} \mathbf{W}_i(x) + \mathbf{N}(x) , \qquad (6.1)$$

where  $\mathbf{S}(x) = [S_1, S_2, ..., S_{N_x}]^T$  is the data,  $\mathbf{W}_i(x) = [W_1, W_2, ..., W_{N_x}]^T$  is the waveform that represents each event *i* and  $\mathbf{N}(x) = [N_1, N_2, ..., N_{N_x}]^T$  is the noise. *k* is the number of events present in the data and  $x = 1, 2, 3, ..., N_x$ , being  $N_x$  the number of channels in the space dimension. This expression is valid for each frequency  $\omega$ . The next step is to build a trajectory matrix from  $\mathbf{S}(x)$ . Up to here, SSA follows the same methodology as the Spectral Matrix filtering proposed by Mari and Glangeaud (1990). The main difference between the two methods is that SSA performs a SVD over a trajectory matrix built from  $S_{\omega}(x)$  and Spectral Matrix Filtering decomposes the covariance matrix  $\mathbf{SS}^H$ , where *H* denotes the conjugate transpose. Mari and Glangeaud (1990) uses the inner product of two events given by:

$$\langle \mathbf{W}_1, \mathbf{W}_2 \rangle = \mathbf{W}_1^H \mathbf{W}_2 , \qquad (6.2)$$

to determine when two events are orthogonal. Two vectors can only be represented by two different singular values if they are orthogonal. Mari and Glangeaud (1990) explain the separation performed by the Spectral Matrix Filtering by analyzing the inner product between each  $\mathbf{S}(x)$  vector of each event. Since SSA decomposes a Hankel matrix, the internal product has to be calculated between the lagged vectors of each event. It is possible to show that when two  $\mathbf{S}(x)$  vectors from two different events are orthogonal, the lagged vectors for the same two events are orthogonal too. This is explained because, as shown in equation 3.5, in the presence of a single event the rank of the Hankel matrix is 1, and its column vectors are parallel to each other, and therefore, they are parallel to the  $\mathbf{S}(x)$  of such event. We can understand from this that all the lagged vectors from a single event will be projected over the same singular vectors. Given that the SVD decomposes the Hankel matrix into its orthonormal basis, it is clear that for each event to be reconstructed by a different singular values, these events have to be orthogonal between each other. It is evident that the number of events to reconstruct has to be lower than the embedding dimension L, named in chapter 2, given that this is the maximum number of singular vectors that represent the data. By computing the inner product between two lagged vectors from the Hankel matrix of two different events, it is possible to know if they are orthonormal. To do so, it is necessary to normalize each vector and perform equation 6.2. If the inner product is near one  $(\langle l_1, l_2 \rangle \approx 1)$  the lagged vectors are almost parallel, if the inner product is near zero  $(\langle l_1, l_2 \rangle \approx 0)$  then the lagged vectors are close to be orthogonal. Given that all the lagged vectors for each event are parallel, if this condition applies to one pair of lagged vectors; it will apply for all the combinations between them. It is important to remember that the orthogonality between the two events is related to its dip and consequently to its velocity, so if the velocities of the events are similar, the lagged vectors are parallel, and if they are different the lagged vectors will be close to orthogonal. Also the amplitude of each event controls if the events can be separated or not in a different way than its orthogonality. Taking as a reference the results from Mari and Glangeaud (1990), we can summarize the conditions in which the amplitude of the events and their orthogonality allows them to be separated:

1. Same Amplitude and Different Velocities  $(\langle l_1, l_2 \rangle = 0)$ :

The first possible condition is when the events present the same amplitudes for all the frequencies, and when their velocities are different, making the lagged vectors between both events at every frequency orthogonal. This means that the inner product between two lagged vectors must be close to zero ( $\langle l_1, l_2 \rangle = 0$ ). Giving that none of the

amplitudes has a preference, the projections over the singular vectors is random at all the frequencies. This means that both waves have the same projection over the first two singular vectors. Figure 6.1 shows a graphic explanation of the situation. In here we present the projection of two lagged vectors over the two singular vectors  $V_1$  and  $V_2$ , but the same representation can be done using  $U_1$  and  $U_2$ . In this situation, it is impossible to separate the two events, and the recoveries from the first and second singular values will present contributions of both events. This was tested in a synthetic record presenting two events with the same frequency content, the same amplitudes and different velocities. The results are shown in figure 6.2. It is evident that the recovered section using the first and second singular values (figures 6.2(b) and 6.2(c)) present components of both events.



Figure 6.1: Same amplitude and different velocities.



Figure 6.2: Result for case 1. a) Input data. b) Event recovered by the first singular value. c) Event recovered by the second singular value. Both waves have the same projection over the first two singular vectors, so they cannot be separated.

2. Same Amplitude and Similar Velocities  $(\langle l_1, l_2 \rangle = 1)$ :

In this case, the amplitudes are still the same, but now the velocities are similar,

meaning that their inner product is close to one  $(\langle l_1, l_2 \rangle = 1)$ . The events are almost parallel, meaning that both of them will be projected in the same singular vector. Because of this, the events will be impossible to separate. This situation is shown in figure 6.3, where we can see how the largest component of both events will be projected over the first singular vector and a smaller component will be projected in the second eigenvector. The example for this case is shown in figure 6.4. Again, figure 6.4(b) is the recovered section by using the first singular value and figure 6.4(c) is the recovered section using only the second singular values. We can see how most of the amplitudes from both events are recovered by the largest singular value while some amplitudes are recovered by the second largest singular value. In case that the events are completely parallel, they will only be recovered by the first singular value.



Figure 6.3: Same amplitudes and similar velocities.



Figure 6.4: Result for case 2. a) Input data. b) Event recovered by the first singular value. c) Event recovered by the second singular value. Both events are partially recovered by the largest singular vector while some amplitudes are recovered by the second largest singular vector. The two events cannot be separated.

3. Different Amplitude and Different Velocities  $(\langle l_1, l_2 \rangle = 0)$ :

The third case is when the events present different amplitudes in every frequency and different apparent velocities, meaning that they are orthogonal (( $\langle l_1, l_2 \rangle = 0$ ). In this case the singular vector weighted by the largest singular value will be aligned with the event with the larger amplitude and the other events will be aligned with the singular vector weighted by the second largest singular value. Given that the events are orthogonal, they will be projected into only one singular vector, which makes possible to separate each of them by recovering only one singular value. In figure 6.5 we can see a schematic representation of this. It is evident that both events are fully projected in each singular vector. An example of this case is shown in figure 6.6. We can see how the section recovered by using the first singular value in each frequency (figure 6.6(b)) contains only the event with the largest amplitude, and the section recovered with the second singular value (figure 6.6(c)) presents only the event with the smaller amplitude.



Figure 6.5: Different amplitudes and different velocities.



Figure 6.6: Result for case 3. a) Input data. b) Event recovered by the first singular value. c) Event recovered by the second singular value. Each singular vector represent each event, meaning that the events can be separated.

#### 4. Different Amplitude and Similar Velocities $(\langle l_1, l_2 \rangle = 1)$ :

The final case is when the events present different amplitudes, but a similar apparent velocity, meaning that the lagged vectors of both events are almost parallel  $(\langle l_1, l_2 \rangle = 1)$ . In here, the singular vector weighted by the largest singular value will get aligned with the event presenting the largest amplitude. Given that the events are not orthogonal, the second event will not be aligned with the second singular vector, resulting in a projection of this event into both singular vectors. In the presence of events with these characteristics, the event with largest amplitude will be fully recovered by the first singular value, but this result will also contain part of the second event. Figure 6.7 presents the graphic representation of this case. It is evident that while the event with the largest amplitudes is fully projected in the first singular vectors.



Figure 6.7: Different amplitudes and similar velocities.



Figure 6.8: Result for case 4. a) Input data. b) Event recovered by the first singular value. c) Event recovered by the second singular value. One event is completely recovered by the first singular vector while the second event is recovered by both singular vectors. In this case the events cannot be separated.

Case	Amplitude	Velocity	Can be separated?
1	=	=	No
2	=	$\neq$	No
3	$\neq$	$\neq$	Yes
4	$\neq$	=	No

Table 6.1: Table summarizing the conditions to separate two events using SSA

The results for these four cases are summarized in table 6.1. From the results of the study of the conditions that allow SSA to separate two linear events we can conclude that this will only be possible when the amplitudes of the events are different and their apparent velocities are also different.

#### 6.3 Methodology

The main objective in these sections is to study the use of SSA to separate GR from reflections in seismic records. It was previously mentioned that the main characteristics of the GR are that it has a low velocity, higher amplitudes and presents lower frequency content than the reflections. Although we can see that both events fulfill the conditions that allow SSA to separate them, regarding amplitudes and velocities, it also presents the problem of these conditions not applying for all the frequencies. These conditions will only apply in the frequency interval where the amplitudes of the GR are larger than the ones from the reflections. If SSA is applied to a frequency band that contains all the frequencies of the GR and the reflections, then the largest singular value will recover the GR in the lower frequencies and the reflections in the higher frequencies. This happens because in the low frequencies, the amplitudes of the GR are higher while in the high frequencies there are only contributions of the reflections. Although the use of SSA separation is not effective to separate GR and reflections from a seismic record with a complete frequency band, it opens promising possibilities when combined with well known processing methods in GR attenuation.

We mentioned before that a common technique to attenuate GR is to use a low-pass filter to model the noise. This method takes advantage of the low frequency nature of the GR. We also mentioned that a drawback of this method is that it also filters low frequency components of the reflections. The use of SSA for separating events arises as a suitable solution to the main problems of using a low-pass filter for GR attenuation. The reason for this comes from the fact that the modelled GR resulting from the filter presents, in general, higher amplitudes than the low frequency components of the reflections that were also recovered by it. If the GR and the low frequency reflections are separated after applying a low-pass filter, the modelled noise will be free of any component of the reflections. This modelled GR resulting from the use of SSA will then be subtracted from the original data.

When SSA is applied to the result of the low-pass filter, the objective is to model the GR as accurate as possible using the first few singular values. Since SSA assumes linear events, the result can be improved if the GR is horizontalized and its lateral correlation is improved. This can be achieved by applying a linear move-out (LMO) to the record and by applying trim statics determined by cross-correlation (Cary and Zhang, 2009). The methodology to use SSA separation to attenuate GR can be summarized in the following five steps:

- 1. Apply a low-pass filter to the data.
- 2. Apply an LMO and static corrections to horizontalize the GR.
- 3. Compute SSA and synthesize the data with the first k singular value. This is an estimate of the GR.
- 4. Remove LMO and static corrections
- 5. Subtract the result from the original data.

#### 6.4 Results and Discussion

#### 6.4.1 Synthetic Data

The methodology proposed in this section was applied to a synthetic gather with two events of different velocity, amplitude and frequency content. Both events were produced using Ricker wavelets with frequencies of 10 Hz and 30 Hz to simulate GR and reflections, respectively.

Figure 6.9 illustrates the application of a low-pass filter, with a trapezoidal window of 0-3-19-22 Hz. This figure displays a) the section with both signals summed, b) and c) each signal independently, d) the result of the filter and e) the subtraction of the recovered signal from the original data. The normalized amplitude spectrum (to the maximum amplitude of the GR spectrum) is also shown. It is evident that the low-pass filter contains low frequency components of the reflection. The amplitude spectrum of the recovered signal (6.9h) shows how the low-pass filter truncates the curve.

Figure 6.10 shows the result of the application of SSA signal separation to the low-pass filtered section. In this frequency band the GR has higher amplitude than the reflections. This allows SSA to recover only the GR using the first singular value and this way separate



Figure 6.9: Application of a low-pass filter, with a trapezoidal window of 0-3-19-22 Hz to a synthetic record. (a) Initial data with two events. (b) Low frequency event representing the GR. (c) High frequency event representing the reflection. (d) GR recovered using a low-pass filter. (e) Filtered data from the subtraction of (d) from (a). (f) Amplitude spectrum of (a). (g) Amplitude spectrum of (b) and (c). (h) Amplitude spectrum of (d) and (e). It is evident that low frequency content from the high frequency signal was recovered in (d) and filtered in (e). It is also evident in (h) that the signal (e) loses frequency content.

the low frequencies of the reflections that were attenuated by the filter. After subtracting the recovered GR to the original record we can see that the amplitude spectrum of the high frequency reflection was almost completely recovered.

#### 6.4.2 Real Data

The method was also applied to a real shot gather with strong GR (figure 6.11), corresponding to shot number 25 from Yilmaz (2001). First, the data was filtered with a low-pass filter in a trapezoidal window of 0-3-15-20 Hz. After applying the low-pass filter, the section contains mostly GR and some low frequency signal from the reflections. A LMO and static correction is applied to this low frequency section to ensure that the method is applied where the GR is present and to improve the linearity of the events. The SSA method is then applied. The first 4 singular values of the Hankel matrix of data were used to recover the GR. We can see that the recovered GR has few components of low frequency part of the reflections is preserved. The result of the SSA filter is also compared to the application of



Figure 6.10: Application of SSA to the result of the low-pass filter. Same notation as figure 6.9. In here (d) is the GR recovered from the first eigenimage of SSA and (e) is the filtered data which results from the subtraction of (d) from (a). We can see here that there are no components of the signal in the recovered section (d). It is also evident in (h) that the signal (e) maintains its frequency content.

an f - k filter (Yilmaz, 2001) to attenuate the GR (figure 6.11d)). One can observe that the SSA result yield to a better signal-to-noise ratio than the f - k filter result. This difference arise from the difficulty of the f - k filter to identify the aliased GR in early times.

#### 6.5 Summary

When a low-pass filter is applied to a seismic section the result contains all the amplitudes from the GR and amplitudes from the low frequency components of the reflections. By using SSA it is possible to differentiate two signals if they have different velocities and amplitudes in the frequency band of analysis. Given that the GR has a different velocity and amplitude than the reflections in a low frequency band, these events can be separated by recovering a low rank signal reconstruction via SSA. The GR, reconstructed after applying SSA over the low-pass filtered section, is then subtracted from the original data to attenuate the coherent noise without affecting the reflections.

The SSA signal separation was tested on synthetic and real data. Results show an improvement in preserving low frequency amplitudes from the reflections compared to standard low pass filtering.



Figure 6.11: Application of SSA for GR filtering on a real record. (a) Original Data (b) Recovered GR. (c) Filtered Data resulting from subtracting (b) from (a). d) GR attenuation using an f - k filter for comparison. This figure presents the result of the application SSA to a real section. We can observe that the amount of signal present in (b) is very small. (c) Represents an improvement in its signal to noise ratio, while most of the low frequencies of the signal have been retained. Also, the amount of GR attenuated by SSA is larger that the one attenuated by the f - k filter.

## CHAPTER 7

# Conclusions and Recommendations

The importance of attenuating different types of noise in seismic processing is of constant interest in the field of geophysics. This thesis has studied the applications of a rank reduction technique called Singular Spectrum Analysis (SSA) for coherent and incoherent noise attenuation. The main goal was to present an overview of this technique, studying its origins and its further applications for seismic data processing, to finally evaluate the possibility of using SSA as an alternative method in seismic noise attenuation. In addition, this thesis has presented an acceleration method for computing the SSA and M-SSA filter based on the randomized SVD. Another novel contribution is the iterative algorithm proposed for data regularization/interpolation and the application of SSA to the coherent noise suppression (ground roll elimination).

SSA has its origins in the field of time series analysis for the study dynamical systems. An overview of these applications was shown in chapter 2. In this chapter, SSA was shown to work in four steps:

- 1. Embedding of the input data. This step consisted of dividing the time series into a series of lagged windows. These vectors were then organized to form a Hankel matrix.
- 2. Decomposition of the Hankel matrix into its singular spectrum using Singular Value Decomposition (SVD).
- 3. Application of a rank reduction of the Hankel matrix. This was achieved by recovering a subset of its singular values.
- 4. Retrieval of the resulting time series by averaging on the anti-diagonals of the rank reduced Hankel matrix.

Although the intention of this chapter was to introduce the early stages of SSA, it can ultimately be used as a reference by different science fields. Emphasis was placed on the use of SSA for time series decomposition and noise attenuation. Despite the decomposition of time series using SSA is an important field of study, the interpretation of the data components depends on the objectives of the study. For this reason, this decomposition was only described, without examining its meaning. Despite this, the examples and references provided can indeed orient an interested reader to the application of SSA for the decomposition of time series. Noise attenuation using SSA was also reviewed and an example was presented, demonstrating its function. In fact, the noise attenuation property of SSA was the main topic of interest in this thesis and was then applied for seismic data processing.

Chapters 3, 4, 5 studied different applications of SSA related to noise suppression and signal reconstruction in seismic processing. In chapter 3 the application of SSA for random noise attenuation was introduced. This technique was applied in the f - x domain and took advantage of the same signal predictability as f - x deconvolution. The application of SSA in this case was similar to the one applied in the analysis of time series, but it required two extra steps to transform the method to the f - x domain and then back into the t - x domain. The basic steps for the application of SSA for random noise attenuation on 2-D seismic records were summarized as follows:

- 1. Application of a Fourier transform to each channel. This converted the data from the t x domain to the f x domain.
- 2. Given that the reflections could be predicted in each frequency, the process of embedding was applied for each frequency unit. This process was analogous to SSA in time series analysis, but in this case the data consisted of each frequency depending on space. For each frequency one Hankel matrix was built.
- 3. The Hankel matrix obtained from the embedding of each frequency was decomposed using SVD.
- 4. The rank of the Hankel matrix was reduced by recovering only a subset of the singular values.
- 5. The data were recovered by applying an averaging on the anti-diagonals of each Hankel matrix.
- 6. After all the frequencies went through the rank reduction process; they were converted back to the t x domain, where the filtered image was obtained.

In general, rank reduction methods present the problem of selecting the appropriate final rank that would lead to satisfactory results. Chapter 3 showed that the selection of the

final rank for the Hankel matrix, in each frequency, depended on the amount of events with different apparent velocities and amplitudes. This represents a significant advantage of SSA over other rank reduction methods. The use of SSA for random noise attenuation was tested on 2-D synthetic and real seismic data. Furthermore, pre-stack and post-stack data was included. These examples compared the results of SSA with the results of f - xdeconvolution. The first example applied SSA and f - x deconvolution on a synthetic gather with linear events and no random noise. Results from this example showed that no signal is affected under these conditions. In addition, chapter 3 presented two examples that applied SSA and f - x deconvolution to synthetic data contaminated by random noise, in addition to, linear and hyperbolic events respectively. Giving that SSA assumes linear events in the data, the example presenting hyperbolic events had to be applied using overlapping windows. Although none of these examples showed a significant difference in the amount of noise attenuated by either of the methods, it was observed that f - x deconvolution filtered also part of the signal while SSA was successful in preserving it untouched.

An advantage of SSA over other noise attenuation methods is that it can be easily expanded to allow for the use of information from several dimensions of a seismic record. This expansion is called Multichannel Singular Spectrum Analysis (MSSA). MSSA was described in chapter 3, and subjected to further research in chapters 4 and 5. The extension of SSA into MSSA is carried out by building a Hankel matrix of each column vector in one dimension. Next, these Hankel matrices were organized as a block Hankel matrix. In other words, SSA was expanded by building a Hankel matrix of Hankel matrices in the embedding step. If the initial data had two spatial dimensions plus time, the technique was considered to be a 2-D MSSA. If the initial data had more than three spatial dimensions plus time, it was considered an N-D MSSA. Chapter 3 expanded on the application of 2-D MSSA, which was tested with a synthetic example. The theory behind N-D MSSA was also treated in chapter 3, but its practical applications and examples were beyond the objective of this work. The synthetic example showed in chapter 3 demonstrating random noise attenuation using MSSA showed a significant improvement over SSA and f - x deconvolution. The amount of noise filtered by MSSA was larger than the other two methods, while the signal remained untouched. The simplicity of this expansion to multiple dimensions and the great improvement produced by it for noise attenuation makes MSSA an important alternative for the attenuation of random noise in seismic records.

MSSA presented a disadvantage compared to other methods of noise attenuation, like f - x deconvolution, which was the large amount of computations required for its applications. This occurred because the block Hankel matrix built in the embedding step increased its size significantly when very few channels were added to the initial data. Given that the SVD was very slow while decomposing large matrices, the rank reduction of a block Hankel matrix took a significant time to run. A solution to this problem was showed in chapter

4 by proposing the application of a randomized algorithm to perform the rank reduction step. This algorithm was proposed by Rokhlin et al. (2009) and consisted of reducing the size of the matrix by means of a randomization process and then of applying two SVD operations. Given that the two SVD operations were applied to smaller matrices the amount of computations decreased significantly. This new algorithm for MSSA using randomized SVD (R-SVD) was tested in several 3-D synthetic records with different sizes and contaminated with random noise. The results showed that the use of R-SVD in MSSA improved its running time by 50%. It was also seen that the outcome of the randomized algorithm lead to results that correctly approximated those obtained by utilizing the traditional SVD algorithm. With this reduction in its running time, MSSA becomes a more useful tool in seismic data processing.

The main challenge of testing MSSA for random noise attenuation in real 3-D seismic records was that it required the data to be regularly sampled in space. This rarely happens in prestack data given that economical and logistical constrains require the displacement of sources and receivers. If the data was organized into a regular grid, some cells did not contain traces. When MSSA was applied over a regularly sampled record, which presented missing traces, it was seen that it could recover part of the amplitudes of the lost signal. Therefore, chapter 5 studied the use of MSSA for the interpolation of missing traces in seismic records. Given that MSSA only recovered a portion of the missing amplitudes when it was applied once, an iterative algorithm was proposed. The latter worked by extracting the recovered traces from one iteration of MSSA and then placing them into the input data of the next iteration. The interpolation using MSSA was tested in a synthetic cube with random missing traces and no added noise, resulting in a very accurate recovery of the missing traces. A second example showed the same synthetic data, but this time contaminated with random noise. The technique was successful to recover the missing traces and to reduce the amount of random noise. Finally, MSSA was applied to a set of 15 CDP gathers whose offsets where irregularly spread. This offsets where regularized into a desired grid, leaving some missing traces. The interpolation using MSSA recovered successfully the missing traces of each CDP. It also attenuated random noise present in the record. In addition, MSSA was applied using the SVD and R-SVD algorithms, providing new evidence of the improvement in running time given by the randomized method.

Chapter 6 introduced a different approach for SSA application. The goal was to attenuate ground roll by separating it from the reflections. For this, a property of SSA that allowed, under certain conditions, to represent different events by recovering individual singular values was studied. This separation was shown to be only possible when the apparent velocity and the amplitudes of the events where different. The separation of ground roll from signal was carried on in a low frequency band, given that in this interval the ground roll presents larger amplitudes than the reflections. The method was tested in a synthetic example, showing that two signals with different frequency content can be separated. It was then applied to a real shot gather, proving to be successful in attenuating the ground roll without affecting the reflections in a significant way. This results shows that SSA can be a very flexible method that can be used for different applications in seismic data processing.

This thesis provided a complete summary of the applications of SSA and MSSA for noise attenuation and data interpolation of seismic records. SSA was shown to be a valid alternative over the use of f - x deconvolution for random noise attenuation. Also, the simplicity of the expansion of SSA into MSSA made the method easy to apply in several dimensions, improving the quality of the results. Problems of large computational times where solved by applying an R-SVD algorithm, which made possible the use of MSSA on larger sets of data. Furthermore, it was shown that applying MSSA in real data made possible the attenuation of random noise, together with the recovery of missing traces. Finally, SSA was applied to attenuate ground roll in seismic shot gather. After analyzing all the applications of SSA for seismic data processing, it is possible to conclude that this method has the potential to become a tool of common use in traditional processing sequences, presenting a good alternative to conventional methods of noise attenuation.

In this thesis I have presented the following:

- A review of the application of SSA for time series analysis, that is helpful to understand SSA and as a reference for future work.
- The connection of Cadzow method with SSA and the relationship between both techniques. The latter is useful to understand the background of both techniques and their equivalency.
- The predictability of the signal in the f x domain and why the Hankel matrix is rank k in the presence of k events. This helps to understand why SSA is successful in attenuating noise, as well as its advantages and disadvantages.
- The application of a randomized algorithm for rank reduction, which decreases the computational time of SSA.
- The application of a method similar to Projection onto convex sets (POCS) that applies SSA iteratively to recover missing traces in seismic records.
- The use of SSA as a technique to separate individual events, which can be applied for ground roll attenuation.

#### 7.1 Future Work

Although this thesis introduced most of the applications of SSA to seismic data processing, some of them can be expanded and improved. For instance, the decomposition of time series using SSA can help to understand the processes that influenced the data. Paleoclimatic records, for example, can be analyzed using this technique (Vautard and Ghil, 1989). Chapter 2 can be used as a starting point for someone interested in applying SSA for the analysis of time series.

It is reasonable to think that if MSSA is extended to more than two dimensions the results would improve. Given that this expansion to N-D MSSA was not tested in this thesis, it can be an interesting topic of further research. This includes its application for random noise attenuation and seismic data interpolation.

Finally, it is evident that the Hankel matrix and the block Hankel matrix present very obvious symmetries. These symmetries can be used to design a rank reduction algorithm that does not need to go over all the elements of the matrix. Such algorithm would require a smaller amount of computations than the traditional SVD algorithm or R-SVD. This would ultimately make MSSA significantly faster to apply, allowing it to work on very large records.

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## APPENDIX A

# Singular Spectrum Analysis Library in Matlab

The codes for the different applications of SSA showed in this thesis were compiled on a Matlab library. Table A.1 shows the different functions of the library. This table indicates the input data supported by each function. 1D data means time series, while 2D and 3D means the dimensions of the seismic input. These codes can be found in *www.saig.ca* 

Function	1D	2D	3D	Description
SSA_X	Yes	No	No	Function for the application of SSA for time series analysis.
SSA_FXY	No	Yes	Yes	Function to apply SSA on seismic records using the traditional algorithm for SVD for the rank reduction step.
SSA_FXY_INTERP	No	Yes	Yes	Function that applies the iterative algorithm to re- cover missing traces on seismic records. It uses the traditional algorithm for SVD for the rank reduction step.
RAND_SVD	No	Yes	No	Function that applies the randomized SVD function described in chapter 4 to perform the rank reduction step.
SSA_FXY_FAST	No	Yes	Yes	Function to apply SSA on seismic records using the RAND_SVD function in the rank reduction step.
SSA_FXY_FAST_INTERP	No	Yes	Yes	Function that applies the iterative algorithm to re- cover missing traces. It uses RAND_SVD function in the rank reduction step.

Table A.1: Table presenting the codes developed in this thesis for the application of SSA.