Improvement of coda phase detectability and reconstruction of global seismic data using frequency–wavenumber methods

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SUMMARY
Due to uneven earthquake source and receiver distributions, our abilities to isolate weak signals from interfering phases and reconstruct missing data are fundamental to improving the resolution of seismic imaging techniques. In this study, we introduce a modified frequency–wavenumber (fk) domain based approach using a ‘Projection Onto Convex Sets’ (POCS) algorithm. POCS takes advantage of the sparsity of the dominating energies of phase arrivals in the fk domain, which enables an effective detection and reconstruction of the weak seismic signals. Moreover, our algorithm utilizes the 2-D Fourier transform to perform noise removal, interpolation and weak-phase extraction. To improve the directional resolution of the reconstructed data, we introduce a band-stop 2-D Fourier filter to remove the energy of unwanted, interfering phases in the fk domain, which significantly increases the robustness of the signal of interest. The effectiveness and benefits of this method are clearly demonstrated using both simulated and actual broadband recordings of PP precursors from an array located in Tanzania.

When used properly, this method could significantly enhance the resolution of weak crust and mantle seismic phases.

Key words: Fourier analysis; Spatial analysis; Time-series analysis; Body waves; Coda waves.

INTRODUCTION
Investigations of the Earth’s internal structures often involve seismic arrays and array-based techniques that shift and stack recordings for different slowness and backazimuth values (e.g. Rost & Thomas 2002). In many applications, vespagrams (Muirhead & Datt 1976; Rost & Thomas 2002) are adopted to identify or detect small amplitude arrivals by using the slowness of a wave as the discriminating factor. Examples include arrivals from the lowermost mantle (e.g. Cobden & Thomas 2013) or P wave underside reflections off mantle transition zone discontinuities that arrive as small amplitude precursors to the PP phase (i.e. PP precursors; e.g. Rost & Weber 2001; Deuss 2009; Saki et al. 2015).

While signal detection is simple in the case of isolated arrivals, the problem becomes much more complex when multiple seismic phases arrive in the same time window. In the case of PP precursors, the strong coda of the diffracted P wave or the reflected wave at the inner core boundary (PKiKP) may interfere with low-amplitude precursors, thereby hampering phase identification and analysis (e.g. Lessing et al. 2015). Despite high signal-to-noise ratios of the main phases, interferences on weak secondary phases such as PP or SS precursors often result in the removal of significant fractions of potentially usable data. The resultant, often mistaken, the absence of underside reflections from seismic discontinuities (e.g. 410 and 660) is conducive to misinterpretations of the temperature and/or mineralogy of the mantle transition zone. Therefore it is critical to minimize the phase interference between strong arrivals and weak signals of interest for a robust phase identification and extraction on a vespagram.

The most commonly adopted approach is stacking (delay-and-sum) due to its simplicity and effectiveness in suppressing incoherent (though not necessarily low-amplitude) arrivals. Recently, more sophisticated resolution enhancement approaches such as Radon transform (Gu & Sacchi 2009; Schultz & Gu 2013), local-slab-stack transform (Ventosa et al. 2012) and slant-stacklet transform (Ventosa & Romanowicz 2015) have been proposed. The local-slab-stack transform was successfully applied to long-period SS precursors (Zheng et al. 2015), which effectively differentiates between seismic arrivals based on moveout or epicentral distance. Other noteworthy methods, which are primarily used in small-scale seismic surveys to reconstruct gaps in data, include minimum weighted norm interpolation (Naghizadeh & Sacchi 2010),
Singular Spectrum Analysis (SSA; Oropesa & Sacchi 2011; Dokht et al. 2017) and least-squares de-noising and interpolation (Naghizadeh 2012). These rank-reduction based approaches transform the data to frequency-space ($f_k$) or frequency–wavenumber ($f-k$) domains to reconstruct missing information, and have been successfully extended to multidimensional seismic analysis. For example, Kreimer et al. (2013) simultaneously remove the noise and reconstruct a 5-D seismic gather using tensor completion in the $f_k$ domain, and Chen et al. (2016) further improves the robustness of 5-D data reconstruction using the damped truncated singular value decomposition. These powerful reconstruction methods, however, have not been well-documented in global seismic studies. Only recently has SSA been utilized to simultaneously remove the random noise and reconstruct missing traces of SS precursors and P-to-S conversions from mantle seismic discontinuities (Dokht et al. 2017).

In this study, we introduce a novel $f-k$ based method for the reconstruction and resolution improvement of broadband seismic array data. Our method takes advantage of the 2-D Fourier transform, which enables an effective detection of sparse dominant energies corresponding to the weak seismic signals of interest in the $f-k$ domain. The following sections will provide the details of this method, which incorporates a number of useful algorithms in the $f-k$ domain for noise removal, interpolation and weak-phase extraction.

**METHOD**

**Data regularization**

One of the prerequisites for 2-D fast Fourier transformation (FFT) is uniform sampling in both temporal and spatial directions. Uniform sampling is not necessarily needed for the discrete Fourier transformation, however, many seismic applications use FFT and therefore require uniform sampling in space and time. The uniform sampling in time is generally fulfilled by digital seismic instruments via a constant data acquisition rate, the latter requirement could be more problematic in practice since seismic receivers, especially those in regional arrays, cannot always be deployed in a uniform or near-uniform pattern. As a result, data regularization, for example, interpolation and resampling, are required prior to data analysis and interpretation. In this study, we introduce two methods to produce a uniform grid where seismic traces are evenly distributed, both of which assume that the structure in the region of study is well-represented by a 1-D model (i.e. the lateral velocity variation is small). Any deviation from this assumption would introduce a travel time deviation and hence a mislocation vector to the data (Krüger & Weber 1992, Jacobit et al. 2013).

One example of non-uniform spatial sampling is shown in Fig. 1(a) for a synthetic seismogram section of three coherent linear arrivals with two different slowness values. In this section, the data sampling is particularly dense at between 100° – 105°, whereas a clear gap is present at the epicentral range from 106° to 110°.

The first method, which we refer to as ‘Static Resampling’, takes the largest interstation spacing and arranges the data on a uniform grid by applying a partial stacking scheme (Fig. 1b). The resampled grid contains the same number of traces at the same aperture as the original data set. To implement this method, the existing traces are binned and repositioned to the new node location (centre of the bin; Fig. 1b), where all data will be stacked to form the new trace if data exist and otherwise the traces will be left empty.

This method assumes a linear moveout of the target phase, defined as a seismic arrival that is influenced by a coda or a different phase. If the sizes of the bins are too large (e.g. greater than 0.5° for the example in Fig. 4), then the seismic arrivals will be shifted in time to correct for the moveout before stacking. The amount of time shift is calculated from the arrival time difference between the reference phase of the original trace (e.g. the first peak of each trace in Fig. 1a) and that of the new (non-empty) trace based on PREM (Dziewonski & Anderson 1981). In our case, we utilize PP as a reference phase to correct for the moveout of PP precursors due to their similar slowness values. Similarly, phases like SS could also be used to correct for SS precursors. A similar approach has been applied to the construction of a common midpoint gather in exploration seismology (Yılmaz 2001). We will discuss problems due to nonlinear moveout below.

An alternative spatial resampling procedure is to redistribute the traces using a moving window, which we refer to as ‘Sliding Window resampling’. Similar to the ‘Static Resampling’ approach, this method partially stacks the entire data for a constant bin size ($b$). Additionally, a sliding window of size $\Delta$ is introduced during the partial stacking procedure to allow for the overlap between adjacent bins. This method essentially applies a moving average to the data, which slightly reduces the spatial resolution but improves the coherency of the nearby traces. Specifically, after populating the first new trace, the bin is shifted by $b - \Delta$ and stacked. In contrast to the ‘Static Resampling’, this method does not limit the number of new traces to the same as the original array, which permits a more flexible choice of the spatial sampling interval. This method assumes a linear moveout of the target phase just in each bin.

**Data reconstruction**

Despite the uniform spatial distribution, our resampled traces can still suffer from large data gaps at certain distances (see 106° – 108° in Figs 1b and c). We use a Projection onto Convex Sets (POCS) algorithm (Abma & Kabir 2006), a compressive sensing based technique (Candes & Romberg 2007), to interpolate the missing traces. The iterative algorithm is similar to the one suggested by Dokht et al. (2017):

\[
d_{i+1} = d_i + SFT^{-1}(\alpha T_i(D_i))
\]

where $d_i$ represents the data matrix (with missing traces) in the time-space ($tx$) domain for the $i$th iteration, and $D_i$ denotes the corresponding $f-k$ spectrum. The operator $S$ represents the sampling matrix (same size as the original data), with values of 1 where data are missing and 0 elsewhere. The operator $FT^{-1}$ represents the inverse Fourier Transform and $T_i$ is the threshold operator, which sets all values of amplitude of $D_i$ below the threshold to 0. The parameter $\alpha$ is a predefined factor that lowers the threshold after each iteration. The initial conditions of eq. (1) are defined by

\[
T_0 = \max(|D_{0i}|)
\]

where $T_0$ and $D_0$ represent the initial threshold operator and Fourier spectrum of the input data $d_0$, respectively.

We explain this process using a synthetic data set with a missing trace (Fig. 2). The data set is transformed into the $f-k$ domain, where the maximum absolute value of the $f-k$ spectrum is picked and defines the first threshold value for the data reconstruction. The $f-k$ amplitude spectrum below this threshold will be muted by setting all values to zero (Fig. 2b). The new $f-k$ domain signal, which contains...
non-zero values only above the threshold, will be transformed back to the $\tau x$ domain. This transformation procedure partially recovers the signal, which represents the dominant frequency and wavenumber of the data set, on the formerly empty trace. This new trace is subsequently reinserted into the original data set, while the rest stays untouched (Fig. 2c).

The output of the first iteration (Fig. 2c) is then used as input for the next iteration, which is subsequently transformed back to the $fk$ domain. The threshold $fk$ amplitude in the second iteration is lowered (i.e. $\alpha$ percent of that of the first iteration), below which all signals are muted again (Fig. 2d). This procedure is repeated, with a gradually decreasing threshold after each iteration, until a predefined number of iterations is reached or the difference between iterations ($d_{i+1} - d_i$) is small enough (e.g. below a predefined cut-off value). In this example the signals (i.e. amplitude and wavelength) in the $\tau x$ domain remain nearly unchanged after six iterations, which is evidence for convergence. The $fk$ spectrum is now broadened by including the signals outside the dominant slopes, which results in increased frequency and wavenumber contents, and therefore the recovery of greater details in the $\tau x$ domain.

This algorithm is dependent on two variables, $\alpha$ as the decreasing factor of the threshold value and the number of iterations. To determine the optimal $\alpha$ and the number of iterations, it is useful to test various combinations of these two parameters and compare the
results of the reconstructed data $d_{\text{rec}}$ with an ideal data set $d_{\text{org}}$ without missing traces. For a quantitative comparison a quality parameter $Q$ is introduced (Kazemi et al. 2016):

$$Q = 10 \log \left( \frac{||d_{\text{org}}||_2^2}{||d_{\text{org}} - d_{\text{rec}}||_2^2} \right).$$  

(3)

The choices of $\alpha$ and the number of iterations largely determine the robustness of the reconstructed signals and the effectiveness of the interpolation process. For instance, lowering the threshold too fast will cause more noise-related energy in the $fk$ spectrum, thereby degrading the quality of the reconstruction (Dokht et al. 2017). To determine the optimal combination of two parameters for our synthetic example, we calculate the quality parameter $Q$ using a grid search (Fig. 3), which shows that convergence is achieved after 6 iterations for a relatively small (~0.55) factor $\alpha$. Our synthetic data sets are calculated using the IRIS Synthetics Engine (IRIS DMC 2015; van Driel et al. 2016) that provides a fast way of generating a reference data set $d_{\text{org}}$ for a given reference model. We used the same synthetic data sets to calculate $d_{\text{org}}$ for our real data examples.

**Design of an $fk$ filter**

The simplest way to modify signals in the $fk$ spectrum is by muting or filtering a wavenumber range. A key consideration is the number of stations $n$, which influences the number of sampled points in the $fk$ domain. A higher $n$ results in a greater wavenumber resolution. Therefore, depending on $n$, the geometry and array aperture, different shapes of the filter may be adopted (Gubbins 2004).

Several candidate filters may be used, most simply a boxcar. Since ripples and overshoots in the signal (i.e. Gibbs oscillations) are introduced by the sharp edges of the filter (Hewitt & Hewitt 1979), it is useful to design the filter with a transfer function $F(f)$ with a certain slope (Yilmaz 2001). For instance, we could apply a linear function

$$F(f) = m \left( f + f_c \right),$$

with the slope $m$, or a Butterworth power function

$$F(f) = \frac{1}{1 + \left( \frac{f}{f_c} \right)^{2a}},$$

(5)

where $f_c$ is the cut-off frequency (or cut-off wavenumber $k_c$ in case of wavenumber filtering) and the exponent $a$ controls the sharpness...
of the cut-off. We show possibilities of different filters tested here in Fig. S1. To create the $f_k$ filter, we calculate the filter functions for each frequency slice of the $f_k$ domain; therefore, generating a 2-D transfer function (as shown in Fig. S1d) centred around a wavenumber of 0. The corner wavenumbers are defined at half of the maximum amplitude of the transfer function.

**Bootstrapping**

To quantify the uncertainties we use the bootstrap algorithm (Efron & Tibshirani 1991). This is conducted by creating $m$ new data sets, containing randomly selected traces from the original data set. The new data sets consist of the same number of traces as the original data set, but with random traces being left out and others being used repeatedly. The standard error $\sigma$ is calculated as

$$\sigma = \left( \frac{\sum_{k=1}^{m} (d_{\text{stacked}} - d_{k})^2}{(m(m-1))} \right)^{1/2}.$$

Here $d_{\text{stacked}}$ is the original data stack and $d_k$ denotes the $k$th randomly drawn stack.

**PROCESSING STEPS**

We process the data in different steps to fulfill the purpose of data filtering. First, we analyse the spatial data distribution and redistribute based on either a ‘sliding window’ or ‘static resampling’ approach. As mentioned above, the ‘static resampling’ scheme assumes the linearity of the target phase’s moveout in the entire data set, which limits the subsequent processing steps such as reconstruction or $f_k$-filtering; the ‘sliding window’ scheme assumes the linearity of target phase in each bin only. Due to the stacking of other phases with potentially nonlinear arrival times, the signal of interest may, to varying degrees, suffer from decreased amplitudes during the reconstruction and filtering process, especially in the event of large spatial gaps in the resampled data set. In step 2, we reconstruct the missing data of the resampled data set using the POCS algorithm. Artefacts in the data such as high amplitude outliers, a shift in arrival times and differing amplitudes of the target phase can influence the result of the reconstruction, which will be discussed in detail below.

In a final step, we apply the $f_k$ filter on the codas or arrivals that interact with the target phase. Because of the nature of the $f_k$ domain, all signals with a linear moveout in the $tx$ domain also show a linear moveout in the $f_k$ domain and share a common origin ($f = 0$ and $k = 0$). Therefore, the $f_k$ filter removes a part of the lowest frequencies of all coherent signals with a linear moveout. Because of the narrow $k$ window, the filter is the most effective on the data in the centre of the array and least effective on data at the edges.

**EXAMPLES**

**POCS reconstruction**

Fig. 4 highlights the performance of the POCS reconstruction. In this example we adopt a synthetic data set containing two arriving phases (marked as $P$ and $PP$) and test different input configurations. From the example in Figs 4(a) we randomly remove up to 30 per cent of the traces and add 20 per cent Gaussian noise to the remaining seismograms (Figs 4b, e and h). We then apply the POCS algorithm to reconstruct the missing traces, the results of which (Figs 4c, f and i) show the reconstruction of two coherent events. The algorithm also reproduces the noise onto the new traces, which is different from the approach of Dokht et al. (2017) where random noise is simultaneously reduced during reconstruction. The difference between reconstructed and original data sets (see Figs 4d, g and j) suggests satisfactory recovery of the main features, but data sets with large trace gaps (e.g. Fig 4h) exhibit considerable clutter after reconstruction (see Fig 4). The maximum permissible noise levels for these configurations are 70 per cent Gaussian noise in case of two missing centre traces (Fig. S2a), 50 per cent for random missing traces (Fig. S2d) and 50 per cent in the presence of a large data gap (Fig. S2g), beyond which the quality of reconstruction is severely degraded. In all these cases the reconstructed traces show less noise and sharper coherent signals than the original data, although the best-fitted $Q$ decreases with increasing noise levels (see Fig. 5).

We further investigate the limits of our POCS reconstruction by performing several tests using synthetic data. Distortions of the arrival time or amplitude of the target phase or high amplitude outliers in the input data set manifest themselves as erroneous signals in the reconstructed data (Fig. 6). After stacking, for example, in a vespagram, these artefacts are no longer observed (Fig. S3). The influence of an interfering signal on the same data sets as in Fig. 6 is shown in Fig. 7. The main difference is the reduced amplitudes in the reconstructed centre traces (see Fig. 7g). The influence of the high amplitude outlier is also apparent in the reconstruction (Figs 7k and l), especially in the neighbouring trace where a high amplitude signal is introduced after reconstruction. These tests show limitations of our reconstruction algorithm. After reconstructing, filtering and stacking, the results of the bootstrap shows that the time-delays introduce an uncertainty as an increase of the standard error and make it harder to identify the precursors (Figs 8a–c). Although the standard error decreases after $f_k$-filtering in each case (Figs 8d–f), the reconstruction and filtering may introduce artefacts on neighbouring traces (Fig. 7).

**$f_k$-filtering**

To assess the performance of the $f_k$ filter further (Fig. 9), we generate a synthetic data gather where a main phase (e.g. $PP$) and its two precursors with 10 per cent amplitudes are introduced (Fig. 9a). This example is inspired by mantle reflections from transition zone
Figure 4. (a) A synthetic data set of a $P$ and $PP$ phase with 20 per cent added Gaussian noise. The waveforms are created with AxiSEM (Nissen Meyer et al. 2014) using realistic station-event geometries and an assumed earthquake source at 20 km depth. The phases of interest are marked by the blue dashed lines. (b) The data set with the two centre traces removed. (c) The reconstructed data using POCS with $\alpha = 0.59$ after five iterations. (d) The difference between the reconstructed and original data sets. (e–g) Similar to (b)–(d) but for a data set with 30 per cent of traces randomly removed. (h–j) Same as (b)–(d) but for a large data gap of around $4^\circ$. 
discontinuities at 410 and 660 km depths (e.g. Shearer & Flanagan 1999; Lessing et al. 2015). The corresponding vespagram shows clear energy clusters corresponding to the accurate slowness values of the three input time domain signals. Since the \( f_k \) transform maps events with the same slowness to the same \( w/k \) ratio, one can extract precursors from those data sets. To illustrate this we introduce an interfering phase (coda) with the same amplitude but different slowness as the main phase in the time window of 200–600 s (Fig. 9b). This long period phase effectively simulates a gradually attenuating signal observed in the \( P \)-wave, diffracted \( P \) (\( P_{diff} \)) wave coda or PKiKP (e.g. Thomas & Billen 2009; Lessing et al. 2014, 2015; Saki et al. 2015). Our result shows strong interference from the coda wave on the first precursor, whereas the second precursor is nearly unaffected (Fig. 9b). The corresponding vespagram of the coda at a lower slowness provides further evidence for the absence of the first precursor and well preserved second precursor. A similar behaviour has been documented by Lessing et al. (2015), where a \( P \) \( P \) precursor strongly interferes with a core reflection.

After applying our Butterworth \( f_k \) filter, the signal of the interfering synthetic coda is suppressed and both precursors are clearly visible (Fig. 9c). It should be noted, however, that the waveforms are altered due to the filtering process: side-lobes are introduced to all phases and the amplitude/waveform is also slightly modified, but the main shape as a zero phase wavelet in all cases is preserved, that is, no phase-shift is introduced. The application of the \( f_k \) filter effectively removes the coda energy on all traces, though the outermost traces in our example still show residual coda energy. The slope of the \( f_k \) filter and the spatial sampling of the data set are mainly responsible for this behaviour. For a larger data set with \( \sim 60 \) traces (Figs 6 and 7), the remaining energy due to the slope is confined exclusively to the outermost traces, which suggests that the spectrum is better resolved. In all cases, this residual coda energy in the outermost traces is suppressed and only the main phase and precursors are visible during the stacking process in the vespagram procedure.

Further we apply our filter to synthetic data sets, using different event depths, to investigate the application of filtering out depth phases such as \( P P \) or \( SS \) that may interfere with our target phase. We calculate synthetic data sets using source depths of 700 km (Fig. 10) that show depth phases of the \( P \) wave that dominate the time window where the precursors arrive. After filtering we find clearer precursory arrivals, for both \( P410P \) and \( P660P \). An additional test using a 500 km depth source shows similar improvement on the precursors after filtering process (Fig. S4).

The \( f_k \) filter is equally effective on real data from an earthquake in the Philippines (1995 May 5) recorded at the Tanzania BB Experiment Array (Fig. 12). This event was used in a search for \( P \) precursors from mantle transition zone discontinuities beneath the Indian Ocean (Reiß et al. 2017). The unequally spaced traces are resampled using the ‘Sliding Window’ method prior to the reconstruction. The resampled and reconstructed data (Fig. 11b) clearly show the arrivals of \( P \) and \( PP \) at a regular spatial interval. The weak arrivals in the time window that contains the precursors (e.g. between 150–350 s) are also well recovered.

Figs 12(a) and (b) shows the \( f_k \) spectrum and vespagram before \( f_k \)-filtering of the data used in Fig. 11. The \( PP \) phase is clearly visible in the vespagram but \( P \) shows a coda of more than 100 s. The \( f_k \) spectrum displays the \( P \) and \( P \) coda energy at wavenumber 0 after aligning on its slowness. There is no evidence for a precursor near the theoretical arrival times but small, diffuse arrivals are detected in the time window surrounding \( PP \). This recording would generally have been rejected due to difficulties in picking arrivals times (e.g. Thomas & Billen 2009; Lessing et al. 2014; Saki et al. 2015). After applying a Butterworth band stop \( f_k \) filter, the region around wavenumber 0, which includes the energies of \( P \) phase and its coda, is muted for all frequencies in the \( f_k \) spectrum (Fig. 12c). The corresponding vespagram (Fig. 12d) shows a clear \( P410P \) arrival at the expected slowness and travel time, even a weak \( P660P \) arrival can be identified in the data. The amplitude difference between \( P410P \) and \( P660P \), which has been reported previously (e.g. Estabrook & Kind 1996; Deuss 2009), can also be seen here but is unrelated to filtering effects.

**DISCUSSION**

A plane wave approximation is commonly used in global seismology, leading to the assumption of travel time linearity in the \( tx \) domain. This assumption is used for most interpolation and reconstruction techniques, especially in active source, near surface seismic studies (Naghizadeh & Sacchi 2010; Oropeza & Sacchi 2011; Naghizadeh 2012). In this study we adopt a POCS algorithm (Stark 1987) for reconstruction, which are highly effective for both exploration and solid Earth seismic data. A known drawback of POCS is an undesirable tendency to interpolate in time direction (Abma & Kabir 2006), that is, during the removal of spatial gaps some well-sampled traces in the \( tx \) domain may be influenced. Due to the linearity of events within in relatively narrow time windows and spatialathers, this problem could be alleviated through partial stacking based on a sliding time window prior to reconstruction. The POCS heavily relies on the distribution of missing data. For an optimal solution, gaps should be as random as possible in all dimensions, though in reality missing data often follow a linear geometry (e.g. due to complex terrain). Large gaps in data sets show clear limits to the POCS reconstruction, where the POCS tends to produce averaging effects similar to smoothing in 3-D seismic velocity tomography (Koroni & Trampert 2015). Despite these limitations, the spatial interpolation from the \( f_k \) filter is generally effective, especially in view that most of the data have been pre-conditioned through resampling (i.e. the moveouts of the target phase such as \( PP \) are retained while other interfering phases are preferentially suppressed).
Figure 6. Results of the POCS reconstruction, when precursor (PC) arrivals are locally delayed by 4 s (a–d). The reconstructed section shows distorted or doubled waveforms at each trace with a delayed signal. (e–h) Results for a linear distance dependent PC amplitude, with its maximum at 3°. The reconstruction shows satisfying results in the centre of the array, but artefacts are clearly visible at both edges of the array. (i–l) Results for a high amplitude outlier in the data set. Again, artefacts and falsely reconstructed signals can be seen at the edge traces of the array. In all test cases, 30 per cent of the traces were randomly removed before reconstruction.
Figure 7. Same as in Fig. 6, but with an additional interfering phase P2.

The series of examples described above show that it is possible to interpolate complex data sets with a high precision. It is possible to redistribute data on a uniform grid for further processing and, because of its simplicity and efficiency, POCS seems to be a powerful tool in data pre-processing. It only depends on two parameters [number of iterations \( i \) and threshold level \( \alpha \) (eq. 1)], and the best combination of those can be examined with a \( Q \) parameter test. Our algorithm is fast, easy to implement/execute, and it is not restricted to the 2-D used in this study. By adding a second spatial dimension provided by a seismic array, it is possible to apply it to...
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Figure 8. Bootstrap results of data shown in Fig. 6. A total of 200 trials are conducted. For each trial, traces are randomly selected and are aligned on the precursor to correct for the moveout before stacking. Bootstrap results of data with (a) a locally delayed arrival of the precursor (c), (b) distance dependent, amplitude varying precursors (Fig. 6g) and (c) a high amplitude outlier (Fig. 6k). (d–f) Same as (a)–(c) but for the \( f_k \) filtered data. The stacking uncertainties (sigma) around the arrival time of the precursor are significantly decreased, but the distribution of uncertainty becomes broader.

It is worth noting that the limitations of the \( f_k \) filter are largely dictated by the distortions in the data sets. For example, the quality of the reconstruction varies depending on travel time delays or amplitude changes. The target signal in the seismic section, for example, strong arrivals, is usually reconstructed effectively while artefacts such as multiples are not. Locally delayed signals tend to increase the uncertainty of the reconstructed and measured signals.

A key problem in seismic data analysis is the interference of weak seismic phases with unwanted neighboring phases or their coda. At different slownesses, these phases should be separate and distinguishable on a vespagram (e.g. Muirhead & Datt 1976; Rost & Thomas 2002). However, the masking of weak phases by interfering signal is possible (e.g. Lessing et al. 2015), as are misinterpretations. We tested the filter on other events with long interfering coda of PKiKP and Pdiff and the process allows a much larger number of events to be used in the analysis. Even \( PP \) precursor phases that were apparently absent before filtering can be recovered after application of the \( f_k \) filter (Reiß et al. 2017).

We showed the application of the \( f_k \) filter to examples of \( PP \) and its precursors that arise from underside reflections off mantle discontinuities. This is only one possible application and many other cases are possible. However, it is critical that the signal of interest shows a sufficiently large slowness difference to that of interfering phases. In the cases of Pdiff and PKiKP, which often exhibit long coda windows (e.g. Lessing et al. 2014, 2015; Saki et al. 2015), the slowness differences are often greater than 2–3 s deg.\(^{-1}\) relative to \( PP \) wave and its precursors. The \( f_k \) filter removes a frequency slice of the \( f_k \)-spectrum, therefore removes the very low frequencies on all phases originating at zero wavenumber and frequency. But this hardly influences the results as we are working with bandpass filtered data which contains minimum low frequency energies. This method would function equally well in the case of \( Ss \) precursor detection. But in some other cases, where the slowness values of the main phase and interfering phases are too similar, the \( f_k \) filter either suppress both phases or results in insufficient suppression. In principle, \( f_k \) filter works for any phase given sufficient spatial resolution from the array data. However, the wavenumber resolution...
must be sufficiently high to distinguish interfering phases from the target phase with similar slowness values.

CONCLUSIONS

We developed and applied an $f/k$ filter to enhance data quality and detect smaller phases, such as PP precursors, in vesagrams. For an application of 2-D FFT the data have to be sampled regularly in space and time. While the time sampling for seismic recordings is usually given, we introduce two methods to re-distribute the recordings of a seismic array onto a regular grid, where the missing traces are filled using a POCS algorithm. We demonstrate the robustness of resampling scheme with synthetic examples and show that the performance of reconstruction is significantly better for data set with...
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Figure 10. Synthetic data set computed using a source at 700 km depth. (a) Seismograms before filtering. The theoretical arrival times are marked by arrows. (b) The fourth root vespagram of (a), where strong P, pP and sP phases interfere with the waveform of the precursors P410P and P660P. (c) The $f_k$ filtered seismograms. (d) Vespagram of (c). Depth phases of pP and sP are removed from the data, while the onset of both precursors is clearly visible.

Figure 11. (a) Teleseismic recording of an earthquake that occurred on 1995 May 5 at 3:53:43.59 UTC, latitude 12.268 N, longitude 125.281E at a depth of 18.3 km in the Philippine Sea recorded at an array in Tanzania. The data are bandpass filtered at corner periods of 15 and 75 s. Theoretical arrivals of P and PP are marked by blue dashed lines. (b) Reconstructed data (red traces) into an equally distributed grid using POCS with $\alpha = 0.92$ and 30 iterations.

randomly or sparsely missing traces than those with large spatial gaps.

The $f_k$ filter cuts out an area of the $f_k$ spectrum and allows to remove energy with certain wavenumbers (i.e. slowness). We apply this to synthetic data that resemble a case of PP wave and its precursors with a second main phase with a long coda that interferes with the signals we want to recover. While one of the precursors is apparently missing in the unfiltered vespagram, the application of the $f_k$ filter allows the identification of the two precursors. Applying the filter to a real data example shows that we are able to improve the identification of PP precursors, thereby allowing an increase in the number of useful events in the data analysis. The $f_k$ filter is not restricted to PP reflections at the underside of mantle discontinuities but applicable also to other data, given that the slowness values of the interfering phases and the phases one wants to recover are sufficiently different.
Figure 12. (a) The $f_k$ spectrum of data presented in Fig. 11. The data have been aligned on the $P$ wave before filtering. (b) fourth root vespagram of the event presented in Fig. 11. Theoretical arrivals of $P$, P660P, P410P and PP are marked in red. The energy of the $P$-coda is visible between 150 and 220 s. (c) The $f_k$ spectrum of the bandstop filtered data with normalized corner wavenumbers of 0.02 and $-0.02$ and an exponent $a = 2$ (eq. 5) of the data presented in Fig. 5(b). The data were aligned on the $P$ phase before filtering. (d) The vespagram of the $f_k$ filtered data. Phases are marked as in (b).

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REFERENCES


**SUPPORTING INFORMATION**

Supplementary data are available at *GJI* online.

**Figure S1.** Amplitude over wavenumber diagrams showing different shapes of filter functions with normalized cut-off wavenumber at $k = -0.2$ and $k = 0.2$. (a) Simple boxcar bandpass filter, (b) tapered bandpass filter, with $m$ indicating the slope at the edges (shown are examples for $m = 1.3, 2, 4$), (c) Example of a Butterworth shaped bandpass filter, with different values for the exponent ($a = 2, 4, 8$) controlling the steepness of the slope. (d) Frequency over wavenumber display of the transfer function of a tapered $f_k$ bandpass filter function with normalized corner wavenumbers of $0.04$ and $-0.04$ and an exponent $a = 4$. The amplitude spectrum $R$ is shown in grey scale. A Butterworth wavenumber filter of identical shape is calculated for each frequency slice.

**Figure S2.** A synthetic data set of a $P$ and $PP$ phases with added Gaussian noise. The waveforms are created with AxiSEM (Nissen Meyer et al., 2014) using realistic station-event geometries and an assumed earthquake source at 20 km depth. The phases of interest are marked by the blue dashed lines. (a) The data set with the two centre traces removed, with 70 per cent added Gaussian noise. (b) The reconstructed data (red traces). (c) The difference between the reconstructed and original data sets. (d–f) Similar to (a)–(c) but with data set with 30 per cent of traces randomly removed and 50 per cent added Gaussian noise. (g,h) Same as (a)–(c) but for a large data gap of around 4° and 50 per cent added Gaussian noise.

**Figure S3.** Fourth root vespagrams of data sets used in Figs 7 and 8, phases are marked with arrows. (a), (d) and (g) show the vespagrams of the seismograms shown in Figs 8(a), (e) and (i), respectively, (b), (e) and (h) show the vespagrams of seismograms after reconstruction and $f_k$ filtering shown in Figs 8(c), (g) and (k), respectively, (c), (f) and (i) show the vespagrams of the undisturbed data, which are shown in Figs 7(a), (e) and (i), respectively.

**Figure S4.** Synthetic data set computed using a source at 500 km depth. (a) Seismograms before filtering. The theoretical arrival times are marked by arrows. (b) The fourth root vespagram of (a), where strong $P$, $Pp$ and $sP$ phases interfere with the waveform of the precursors $P410P$ and $P660P$. (c) The $f_k$ filtered seismograms. (d) Vespagram of (c). Depth phases of $Pp$ and $sP$ are removed from the data, while the onset of both precursors is clearly visible.

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