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Using prior support information in approximate message passing algorithm

Navid Ghadermarzy

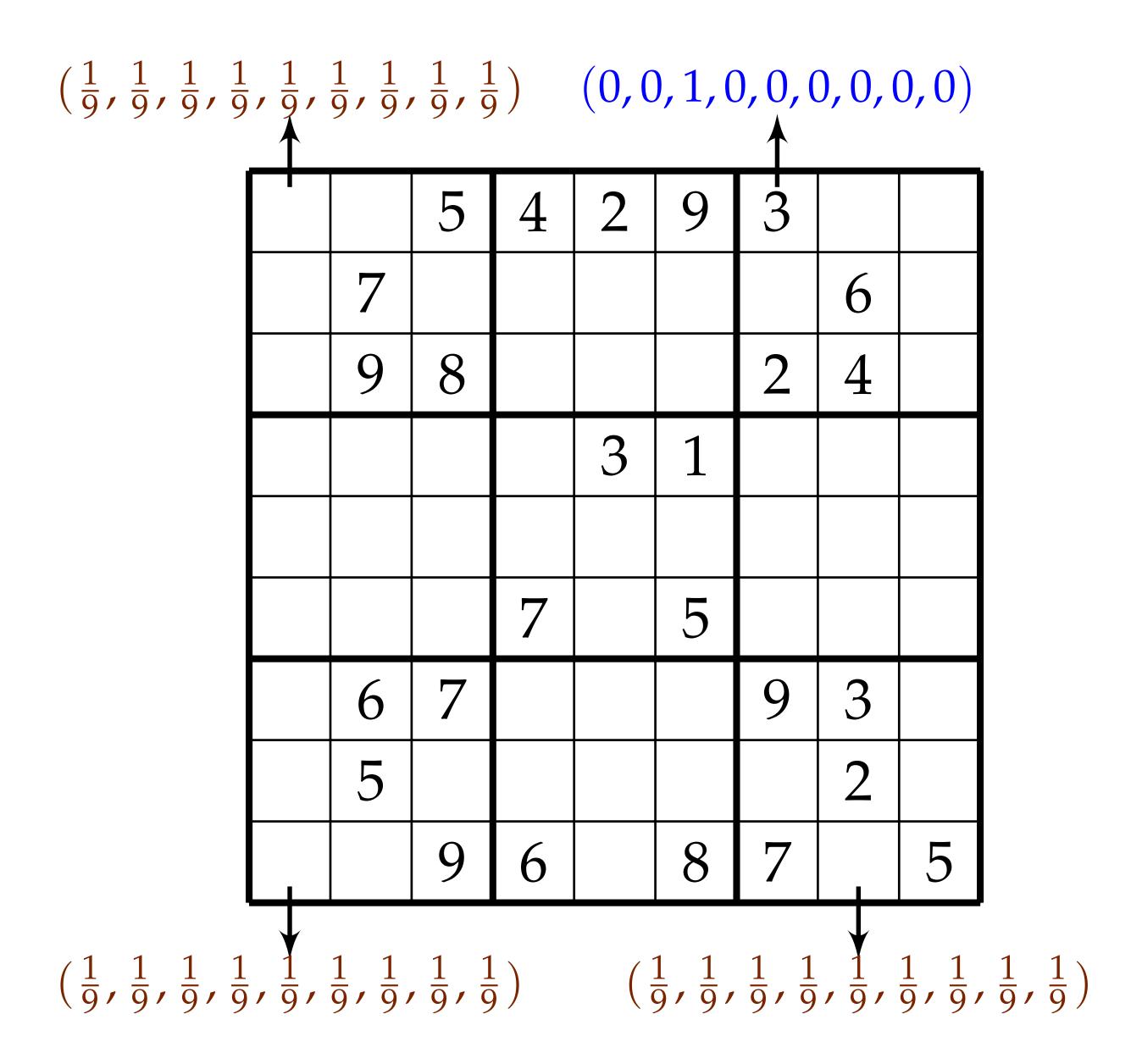
University of British Columbia

December 2, 2013

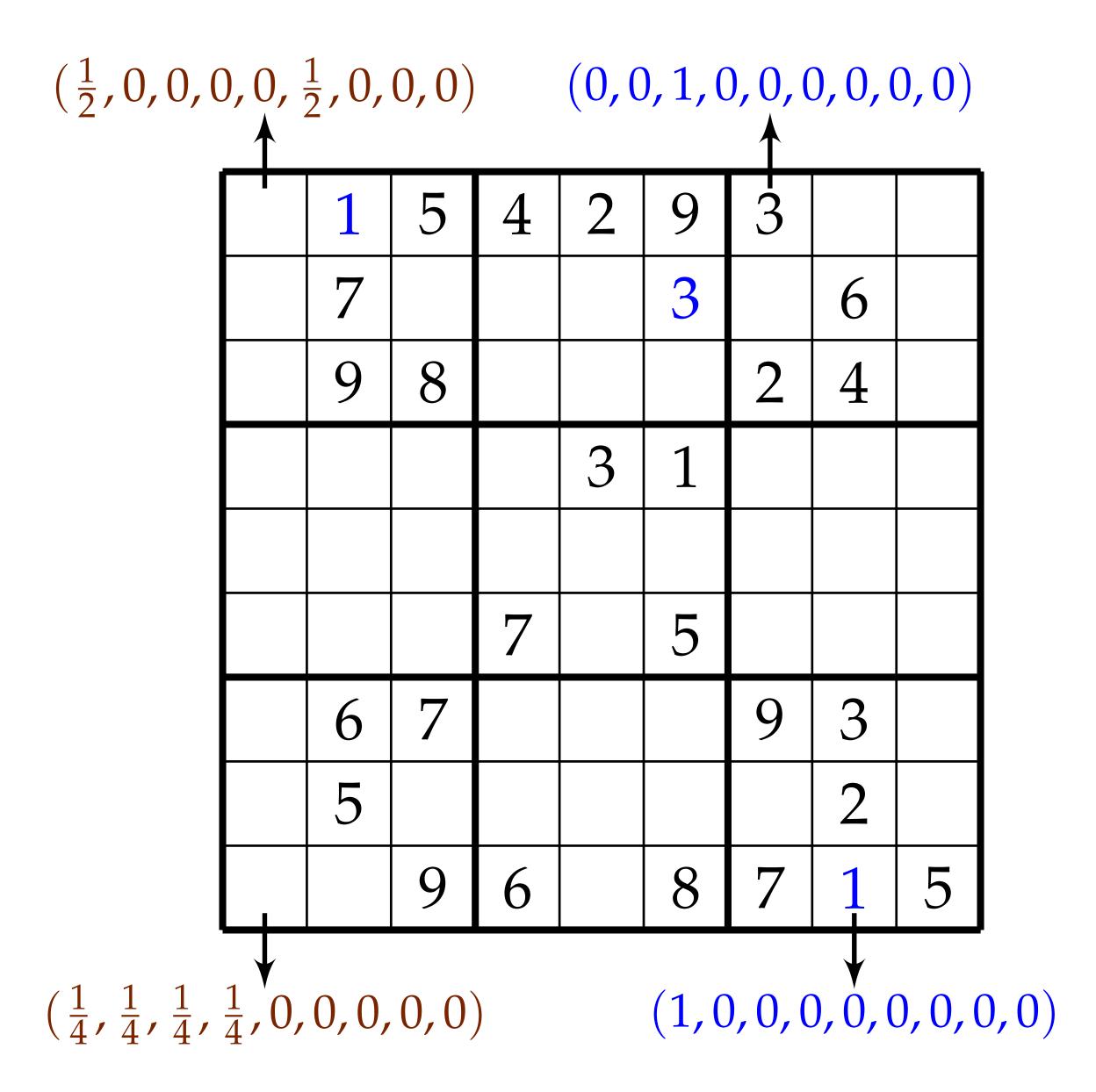
Main messages

- We incorporate prior support information into the approximate message passing (AMP)
 algorithm which is a fast iterative algorithm for sparse recovery.
- Using weighted AMP as a pre-processor we can improve the results of seismic trace interpolation via ℓ_1 recovery in terms of both accuracy and convergence time.

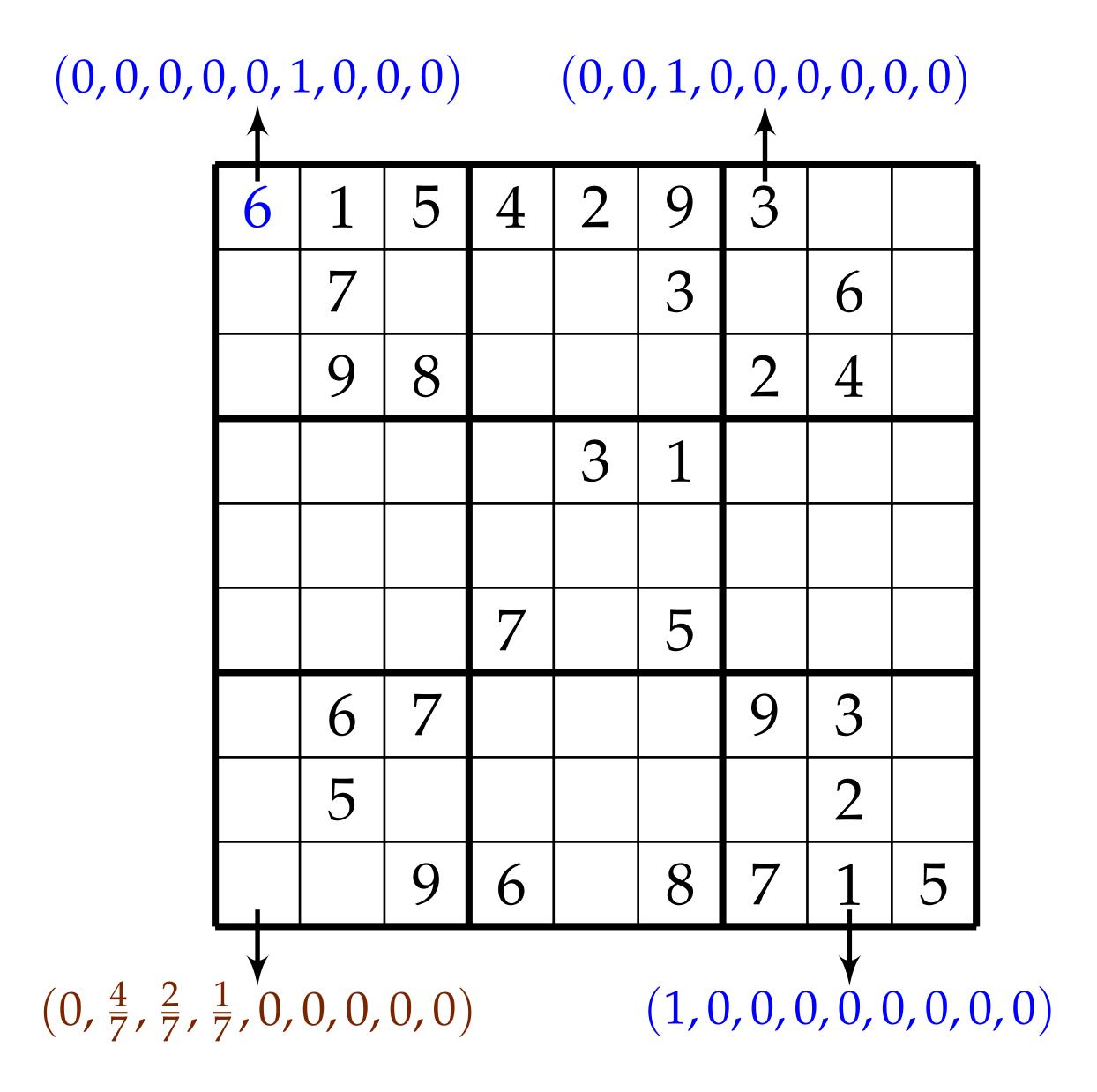
A simple example



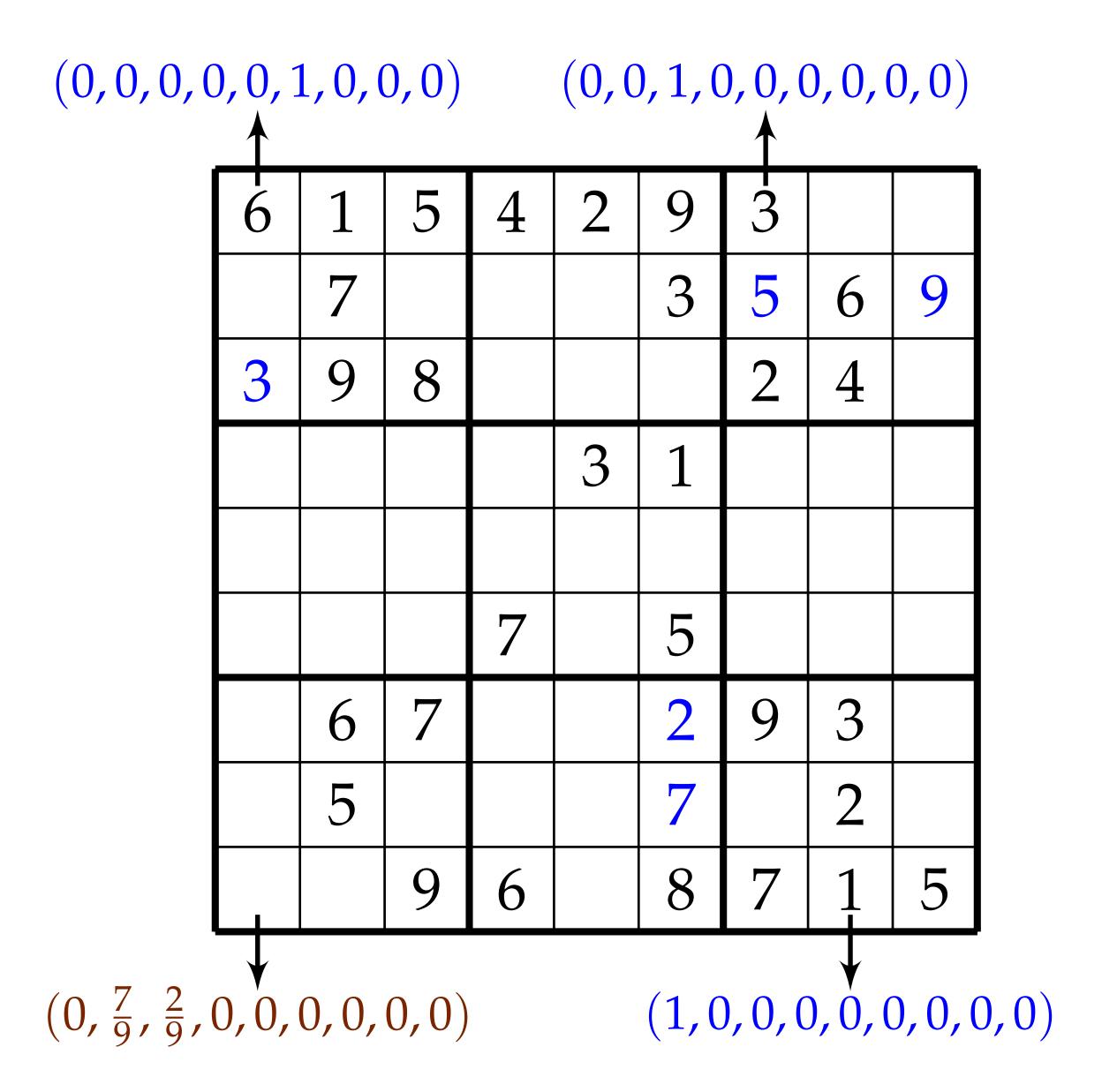
First iteration



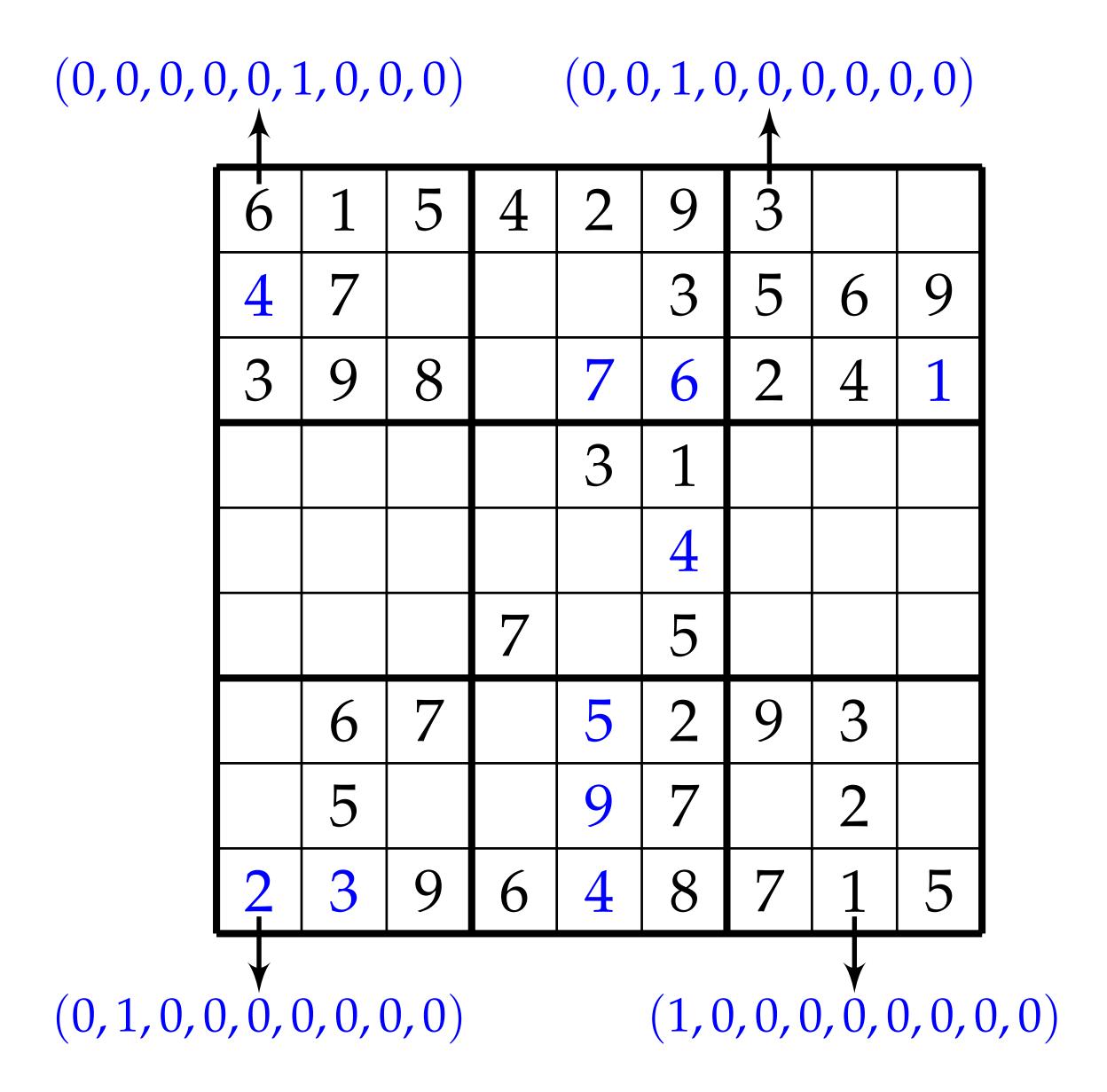
Second iteration



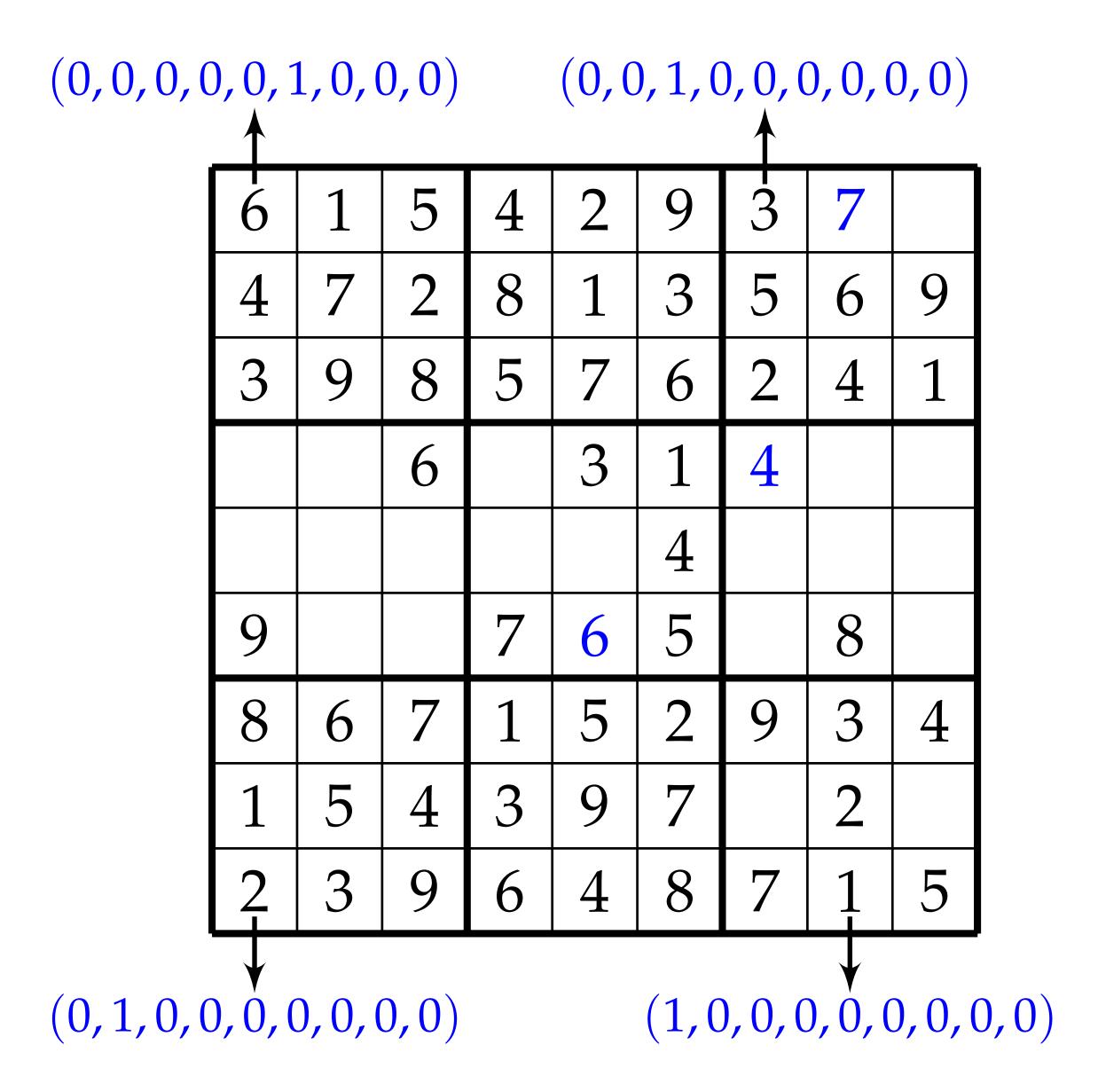
Third iteration



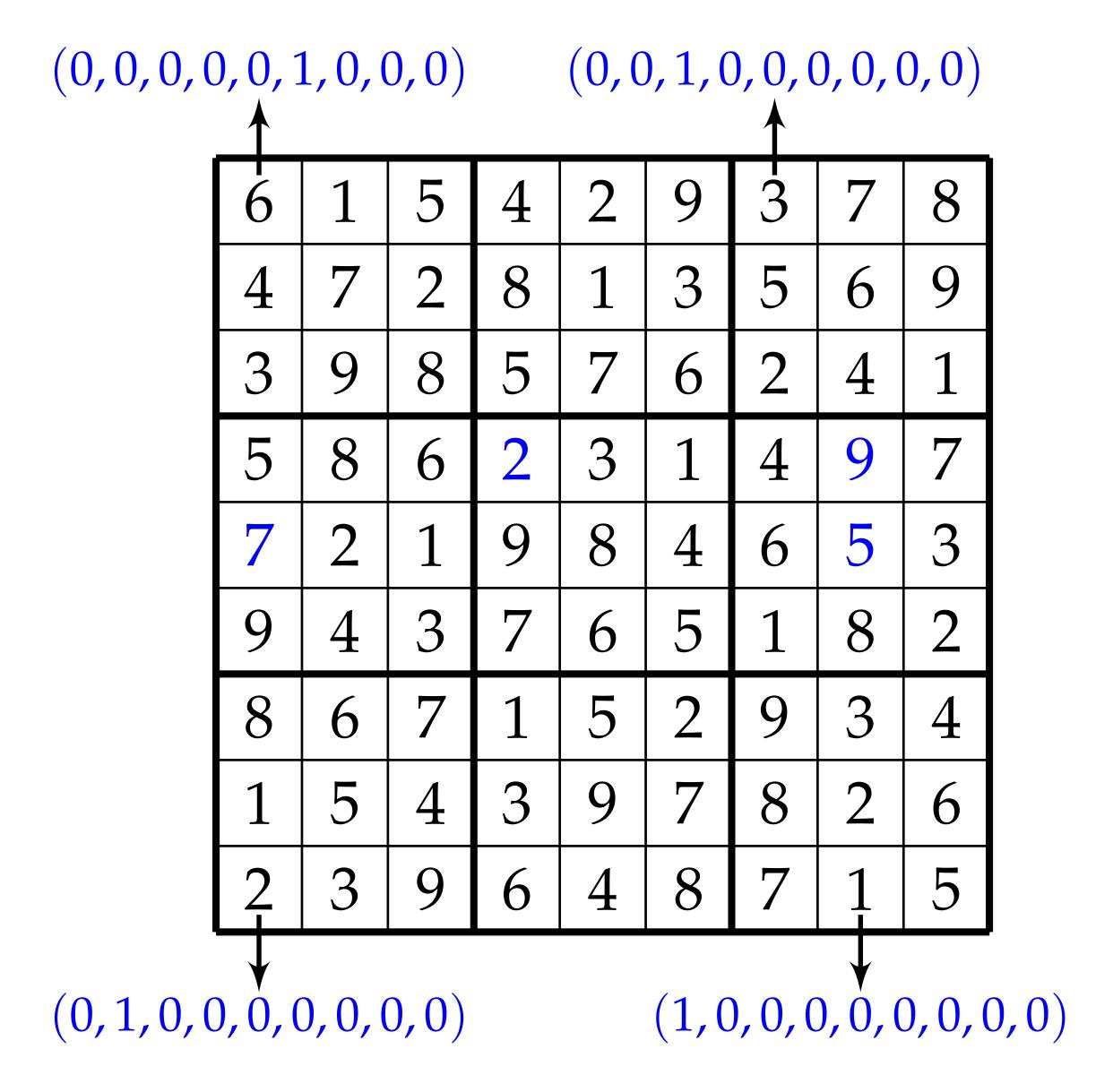
Fourth iteration



8th iteration



12th iteration



Example: Revisiting ℓ_1 for randomized acquisition of seismic lines

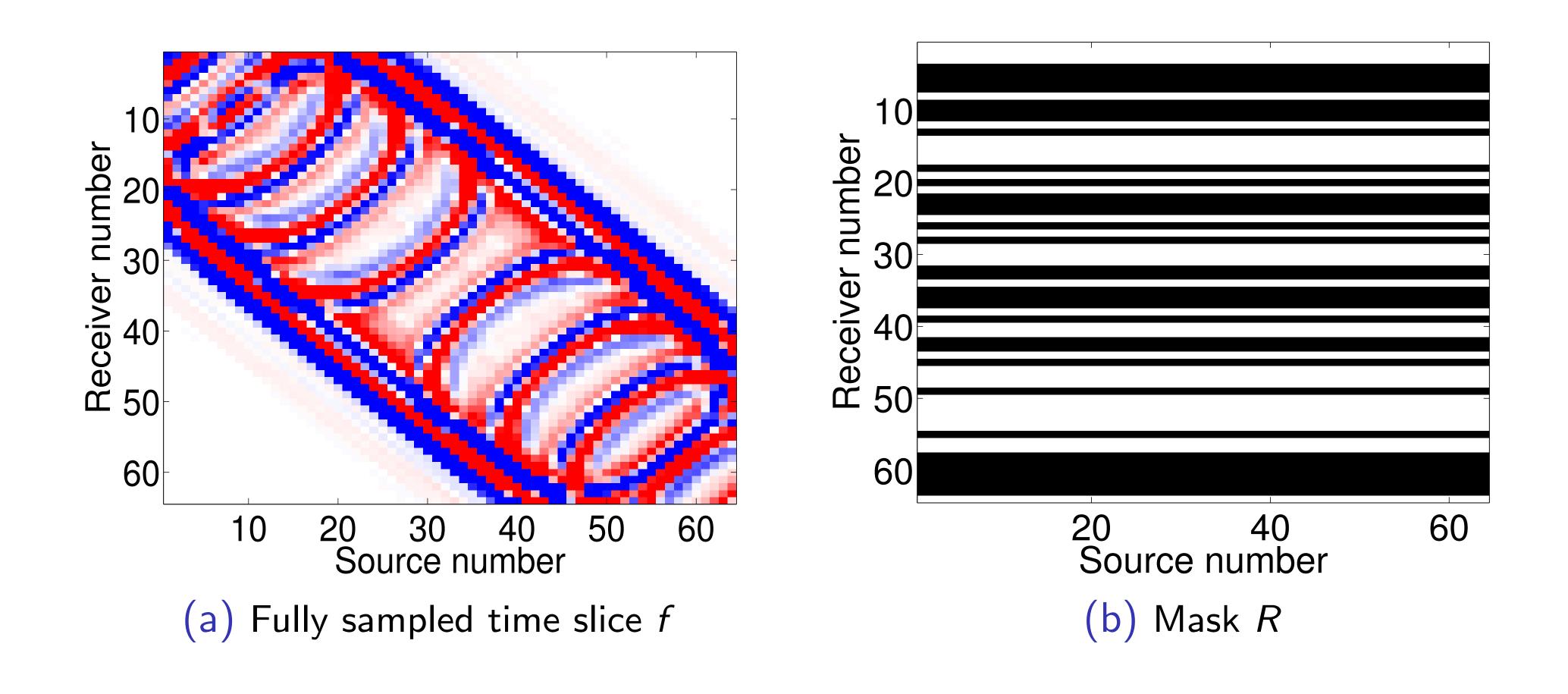
Consider a seismic line with 64 sources, 64 receivers, and 256 time samples.

The receiver spread is randomly subsampled using the mask R.

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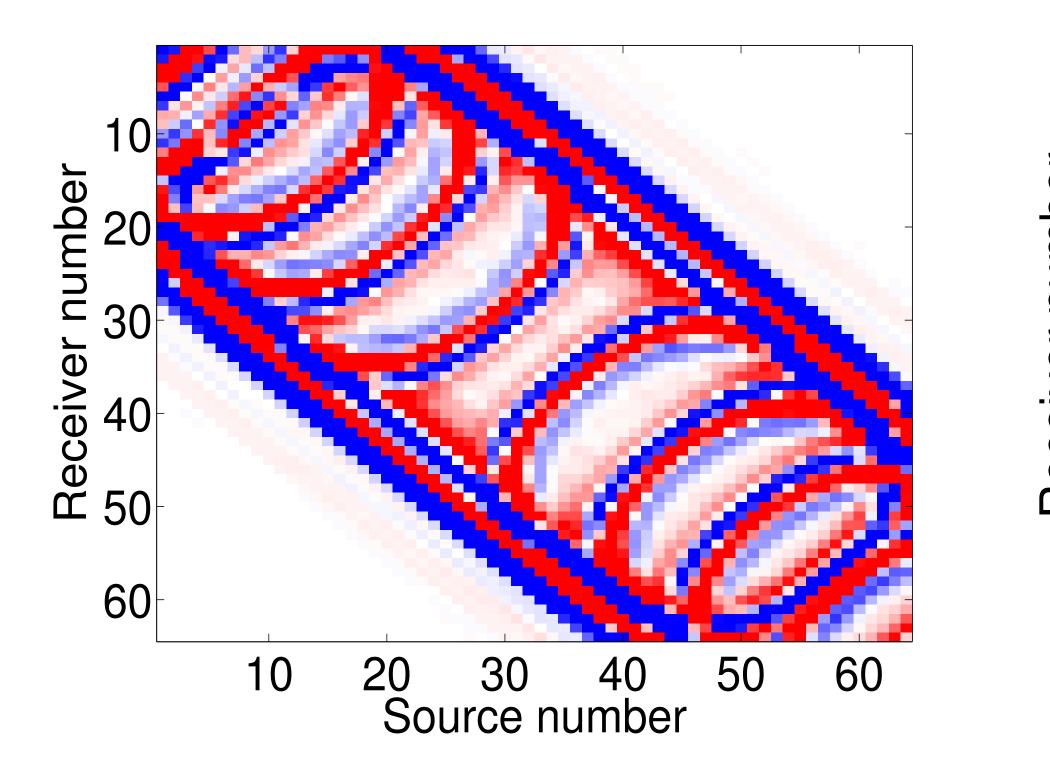
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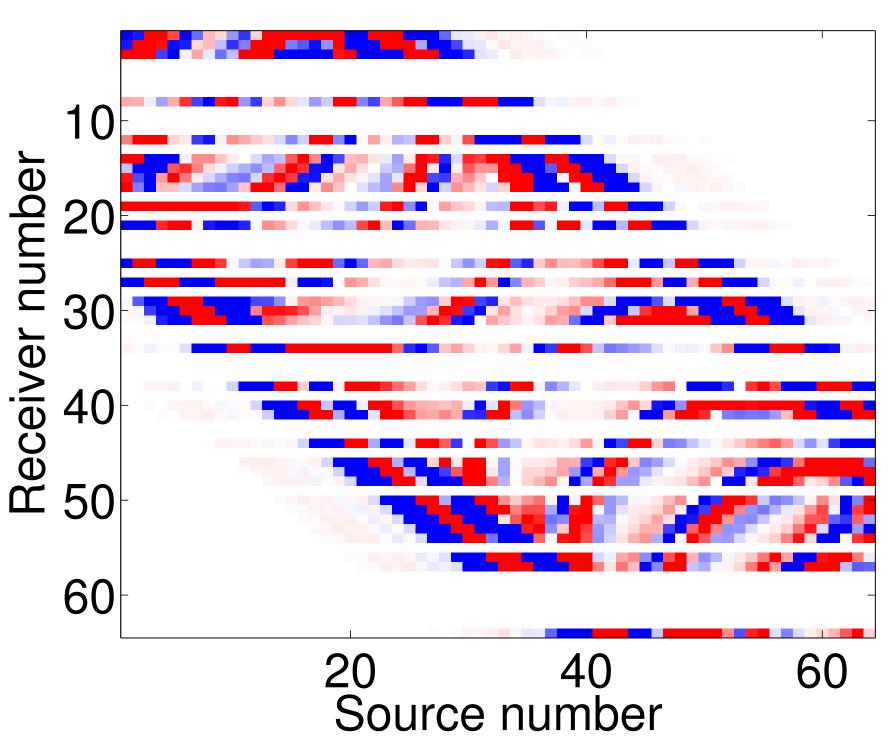


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We want to recover a high dimensional seismic data volume f by interpolating between a smaller number of measurements b = RMf.

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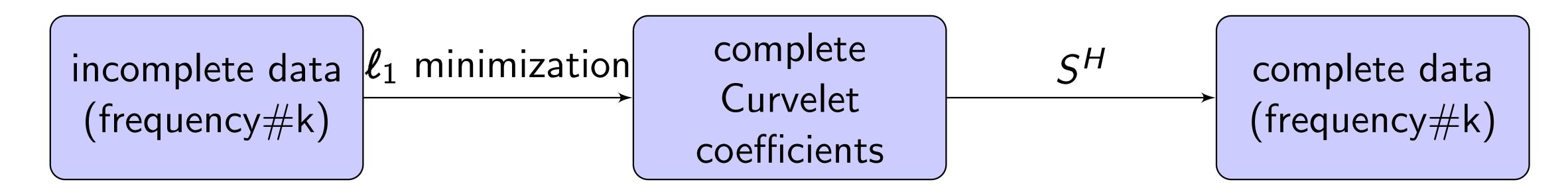
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To recover f from the measurements $b = RMS^H x$, we solve the ℓ_1 minimization problem

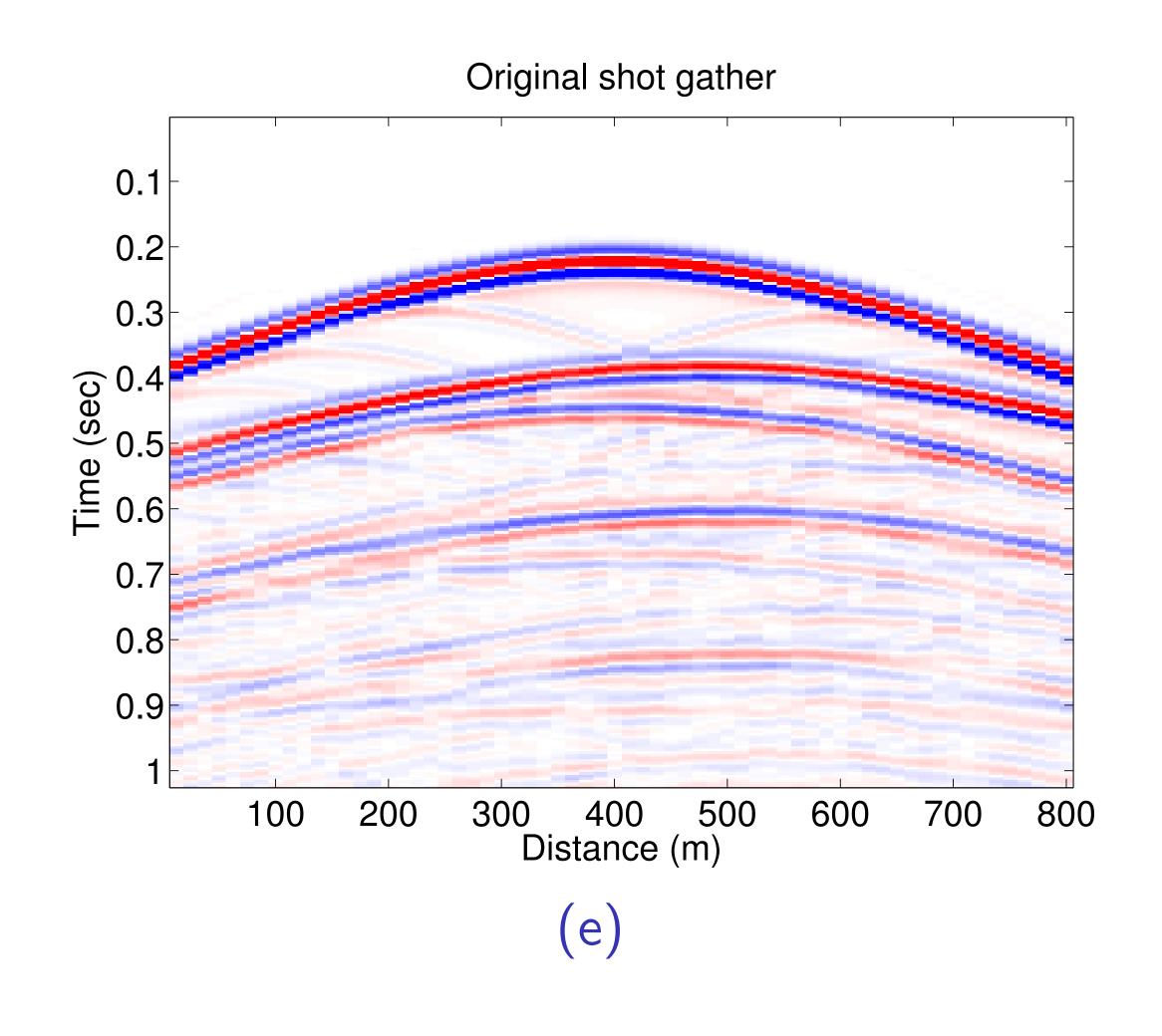
$$x^{\ell_1} := \underset{z \in R^P}{\text{minimize}} ||z||_1$$
 subject to $||RMS^H z - b||_2 \le \epsilon$,

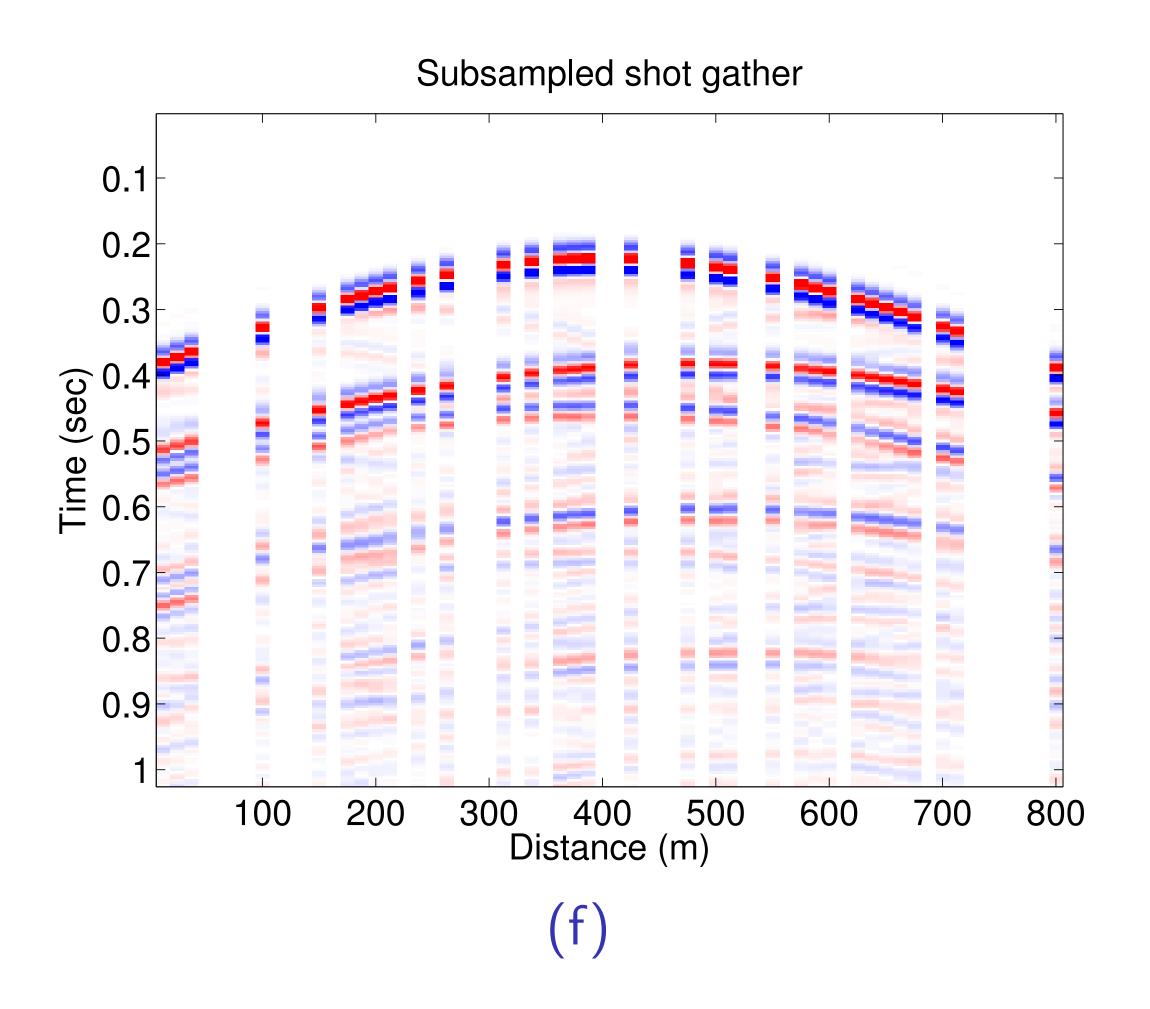
and approximate f by $S^H x^{\ell_1}$.



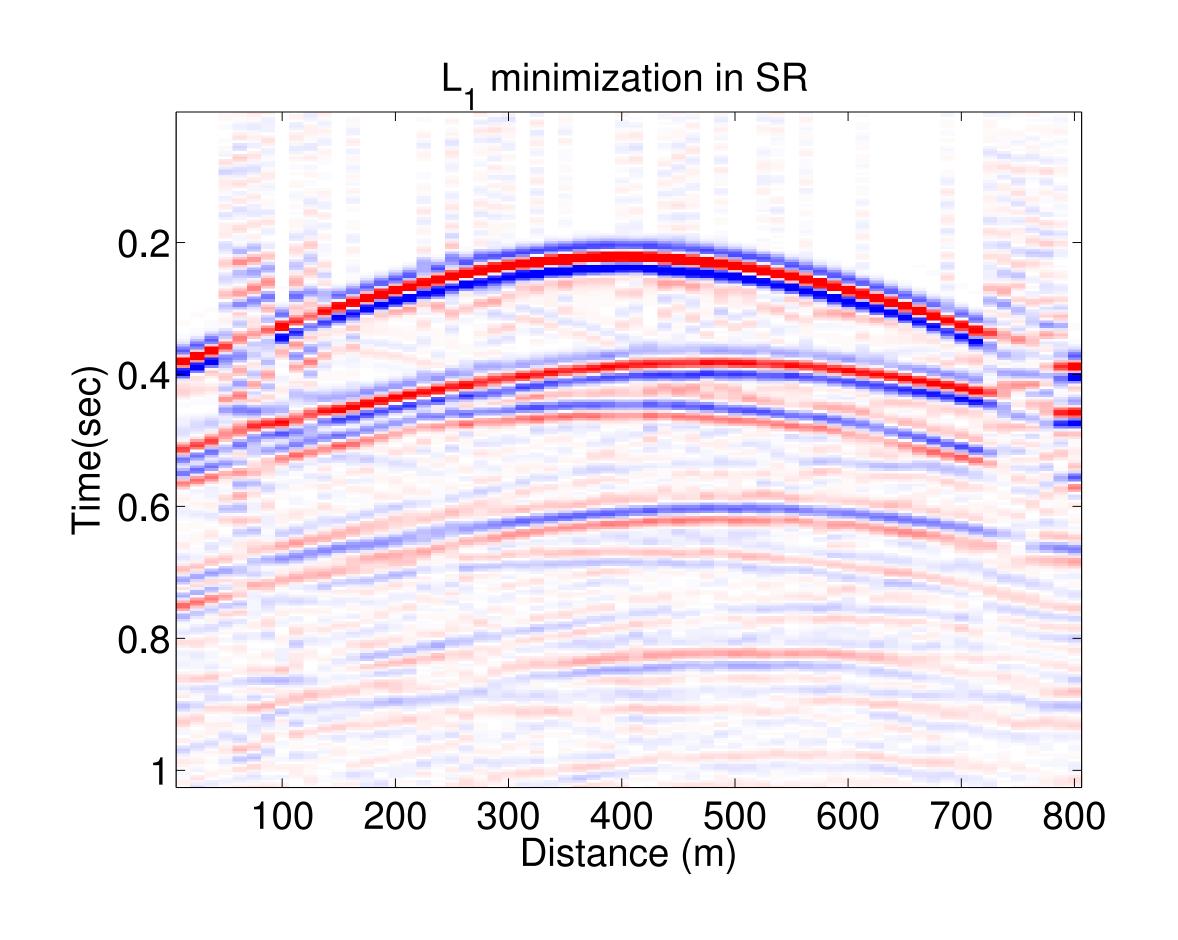
Recovery using ℓ_1 minimization on frequency slices (shotgather # 32)

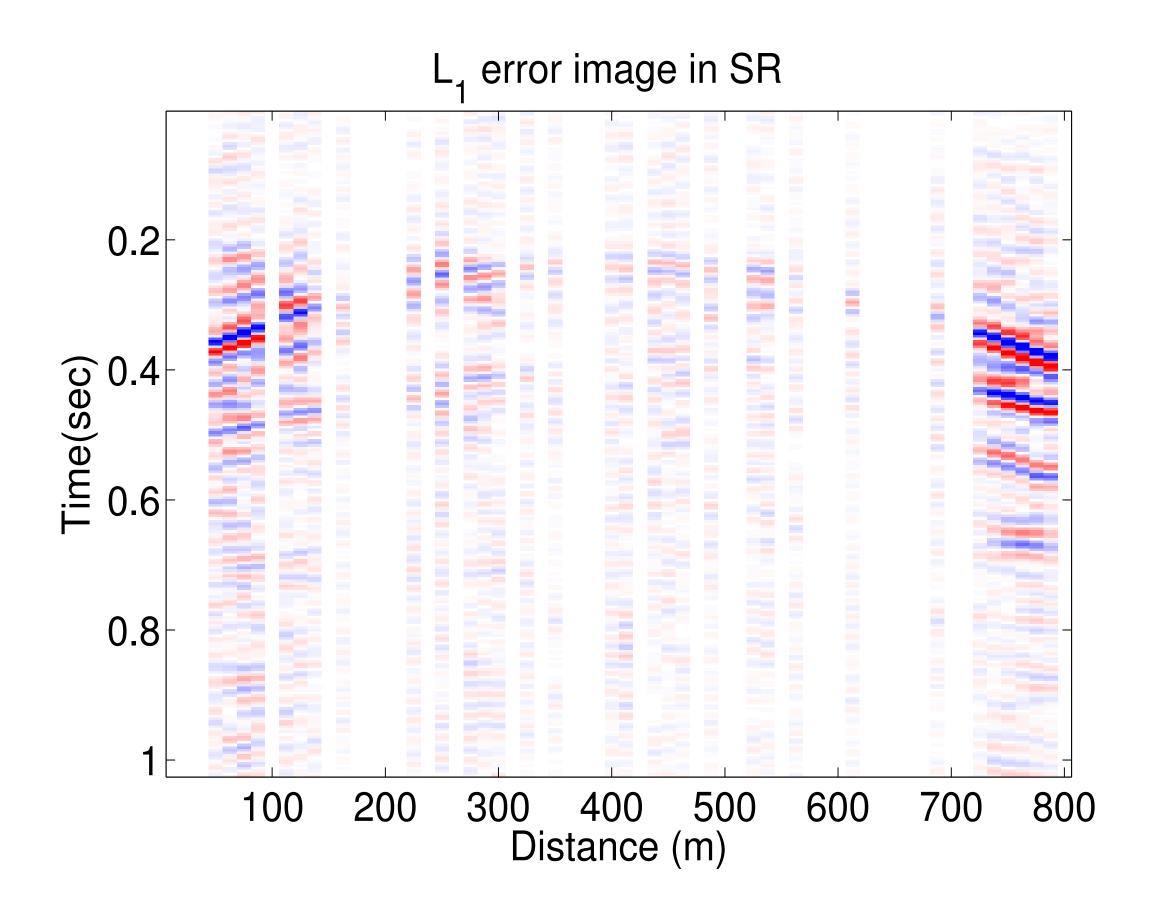
Shotgather number 32 from the seismic line:



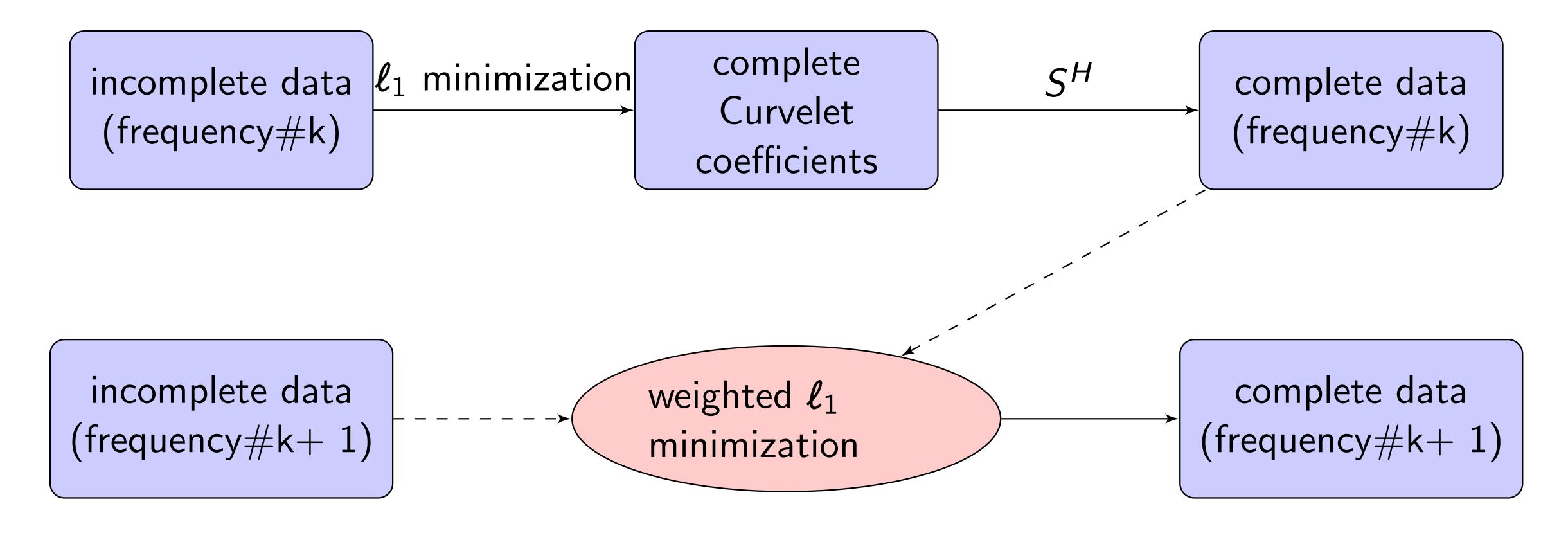


Recovery results: ℓ_1 minimization



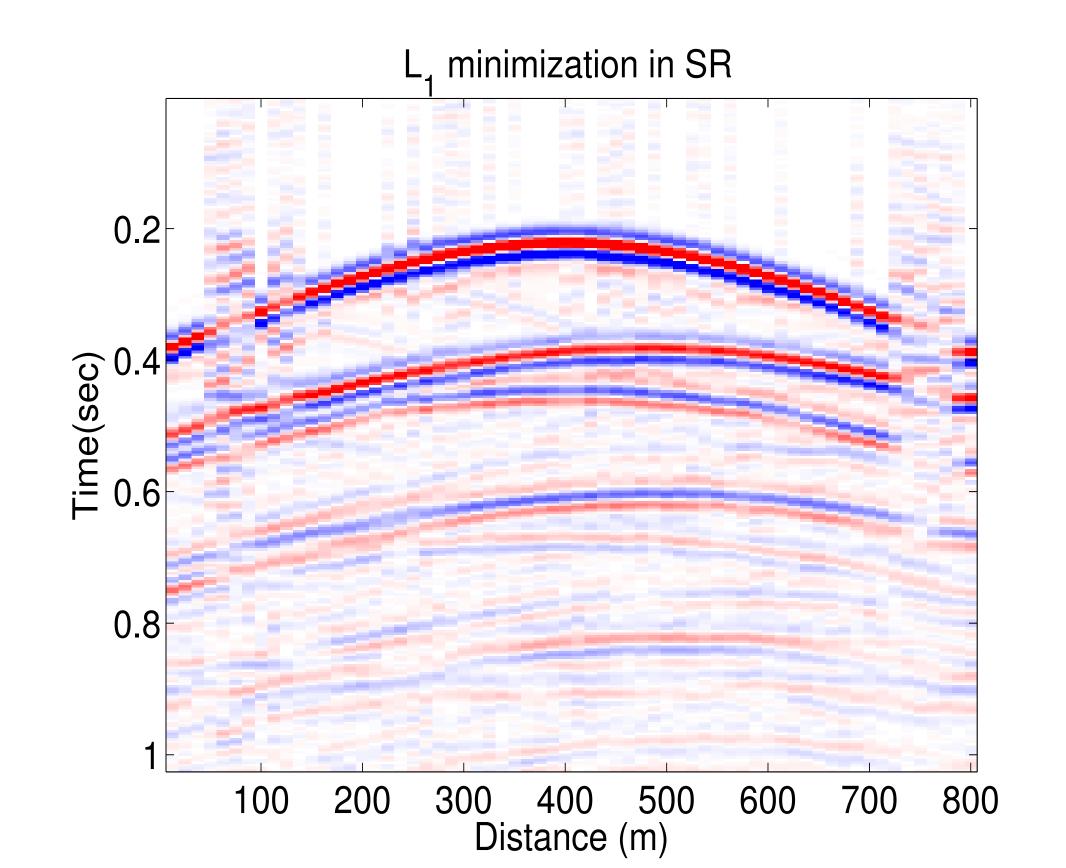


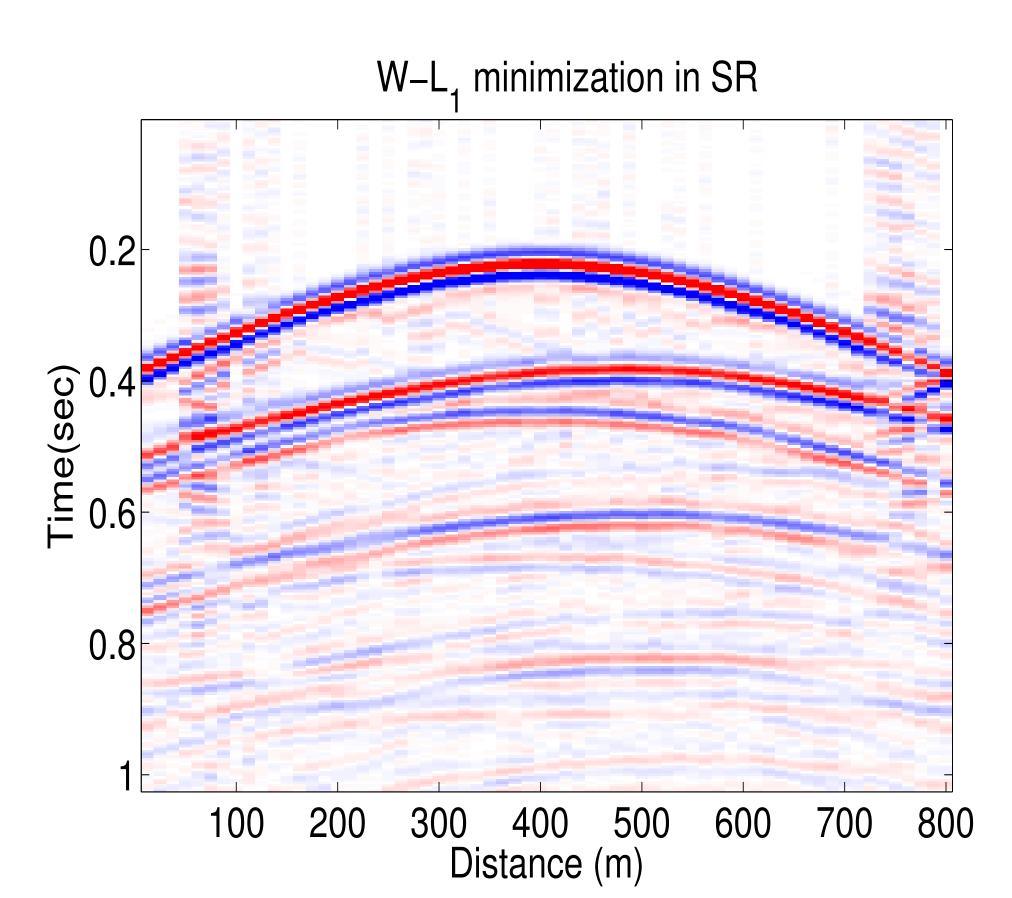
Adjacent frequency slices and have highly correlated curvelet domain support sets. Hence we can use the support set of each slice as an estimate for the support of the next slice.



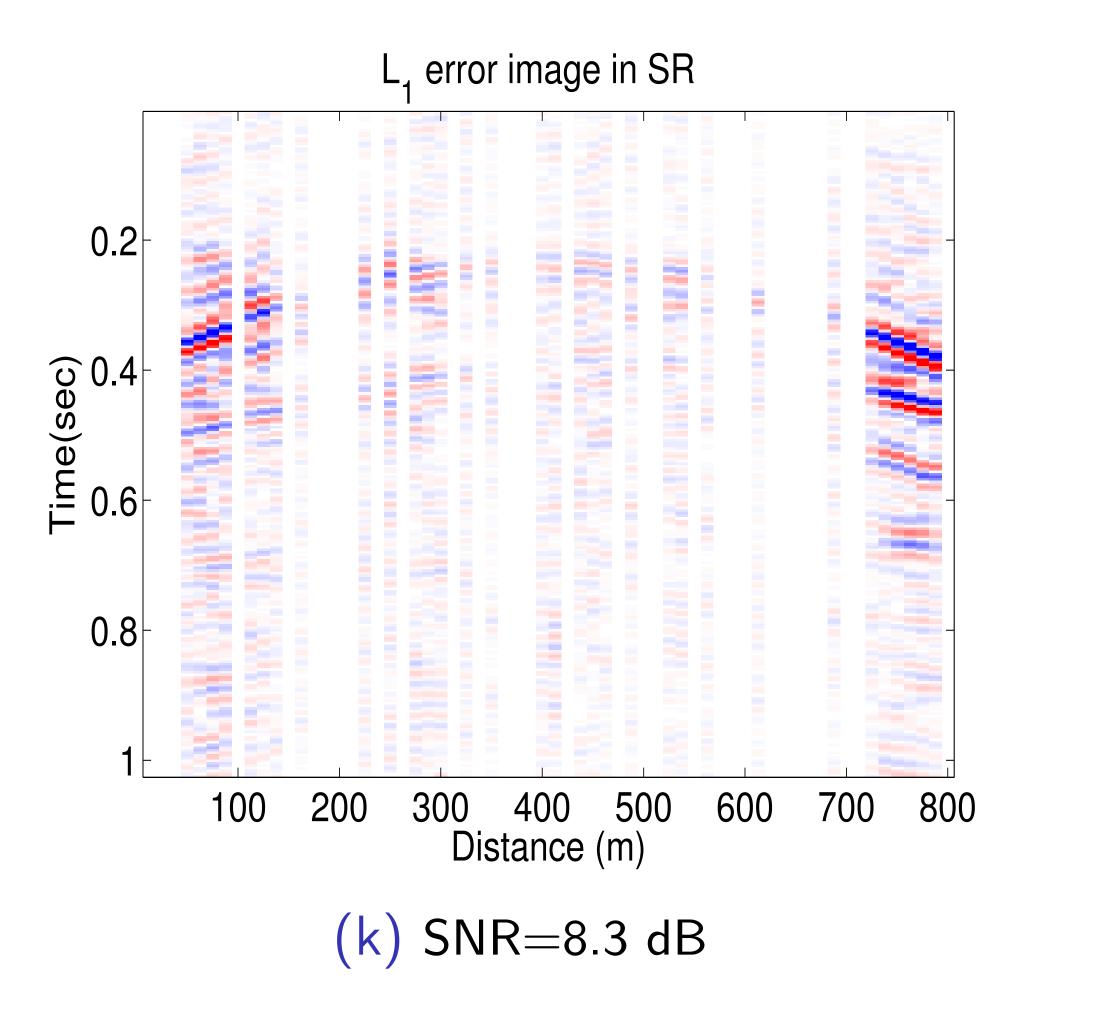
 $x^{w\ell_1} := \underset{z \in R^P}{\operatorname{minimize}} ||z||_{1,w}$ subject to $||RMS^H z - b||_2 \le \epsilon$,

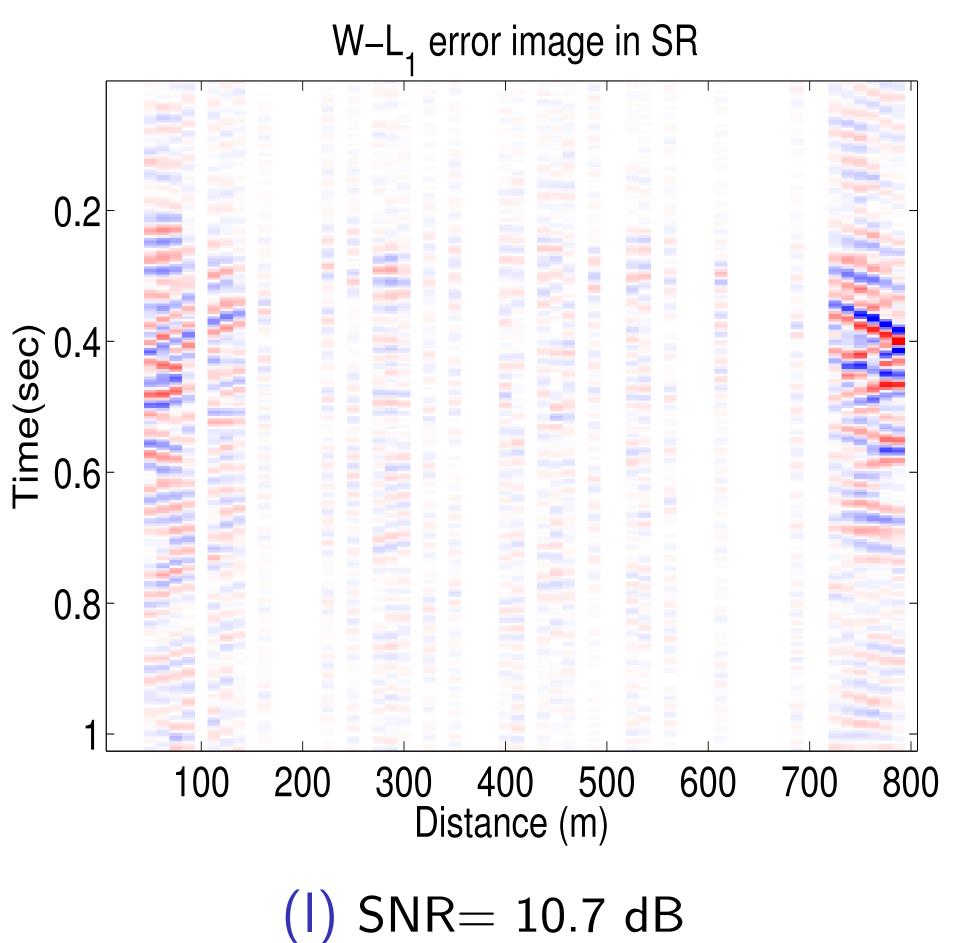
 $\mathbf{w} \in \{\omega, 1\}^N$ is the weight vector and $\|z\|_{1,\mathbf{w}} := \Sigma_i \mathbf{w}_i |z_i|$ is the weighted ℓ_1 norm.





Recovery error: ℓ_1 vs weighted ℓ_1





Iterative thresholding algorithms

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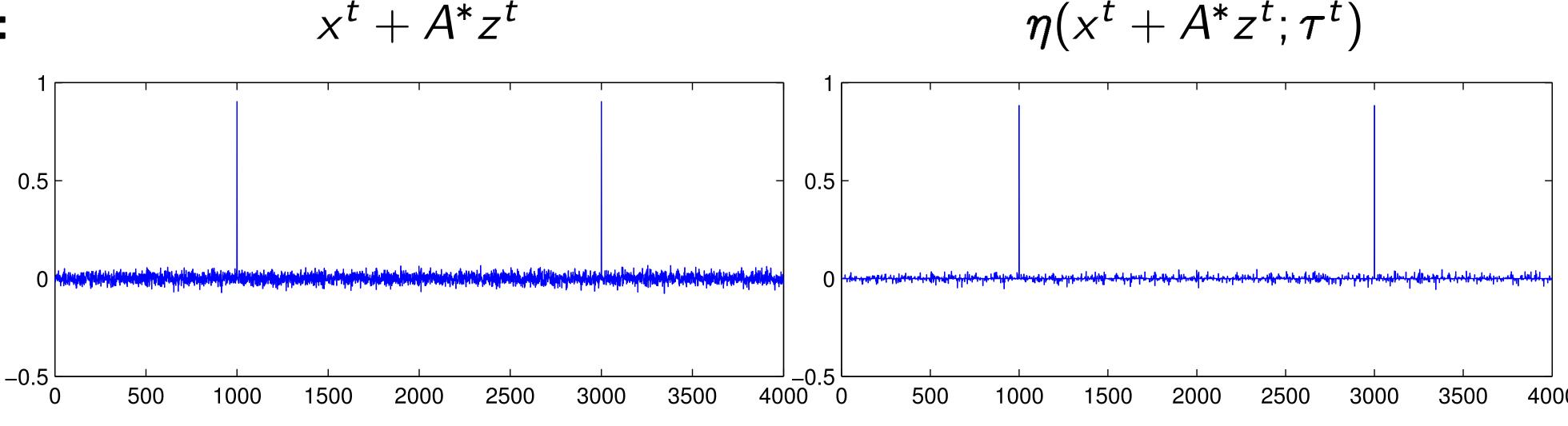
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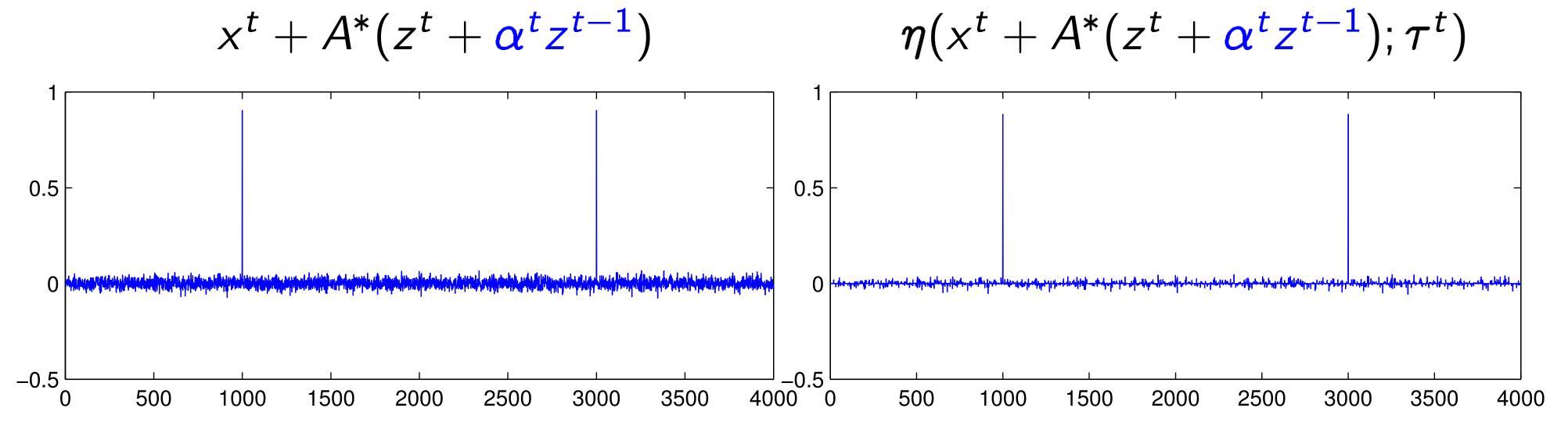
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Simple example with a 2-sparse signali Iteration t=1



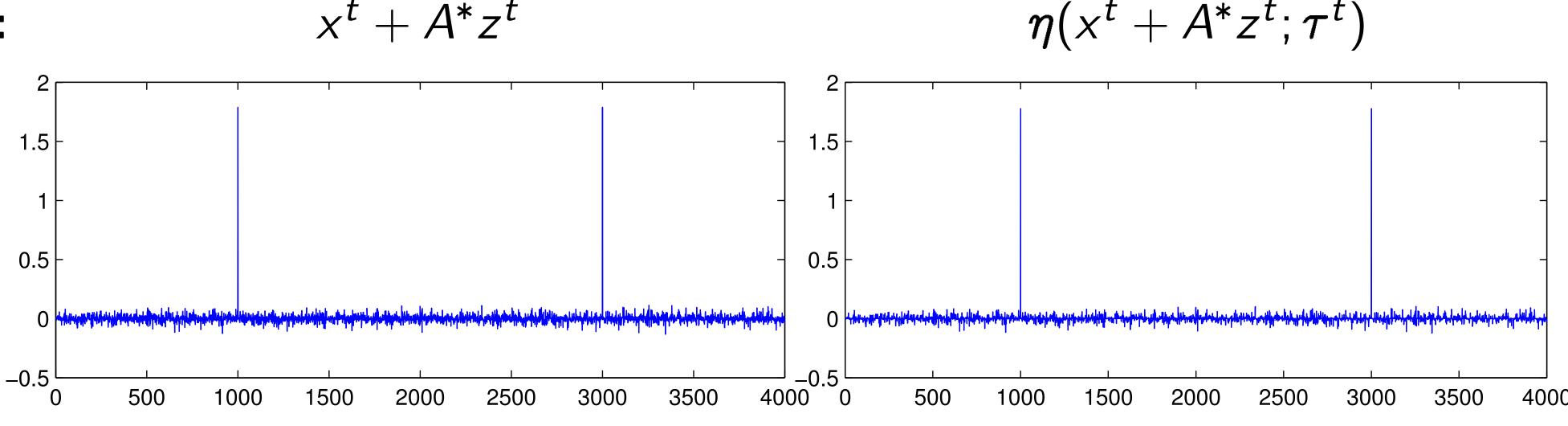


AMP:



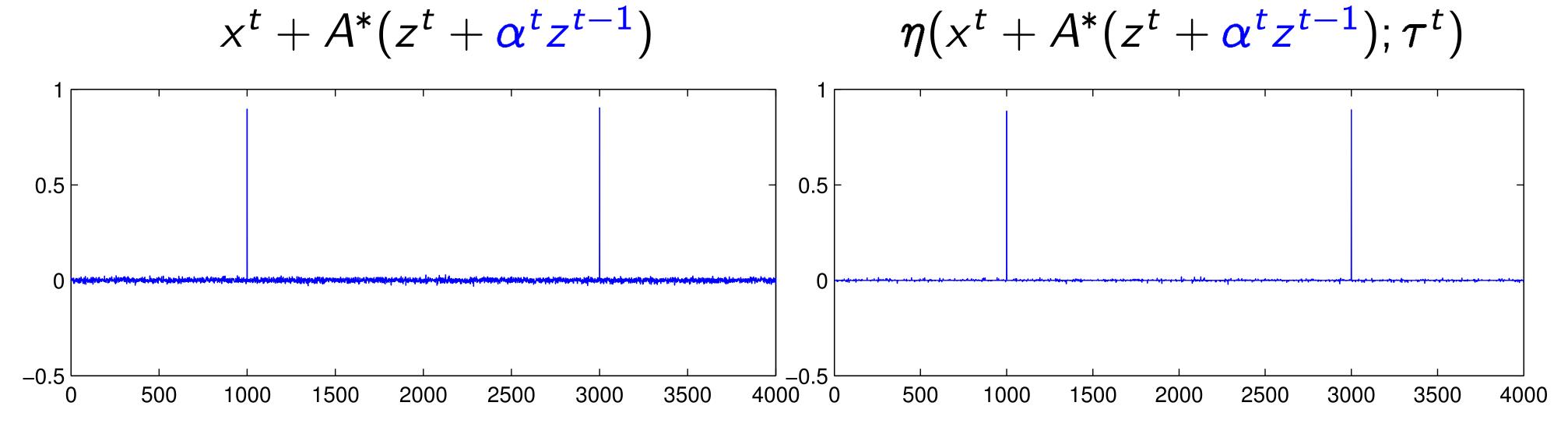
Iteration t=2





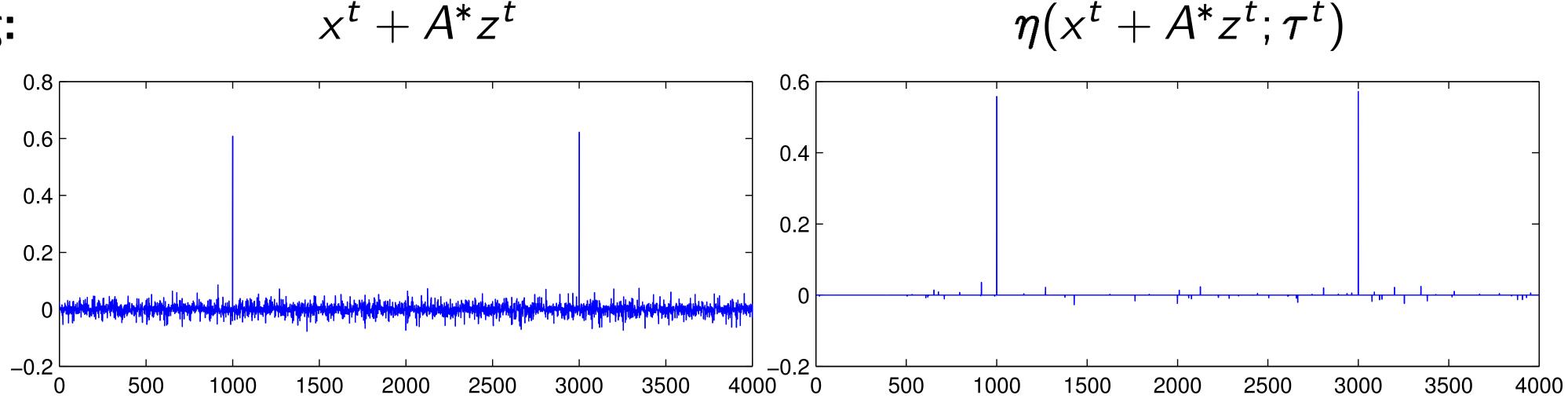
 $x^t + A^*z^t$

AMP:

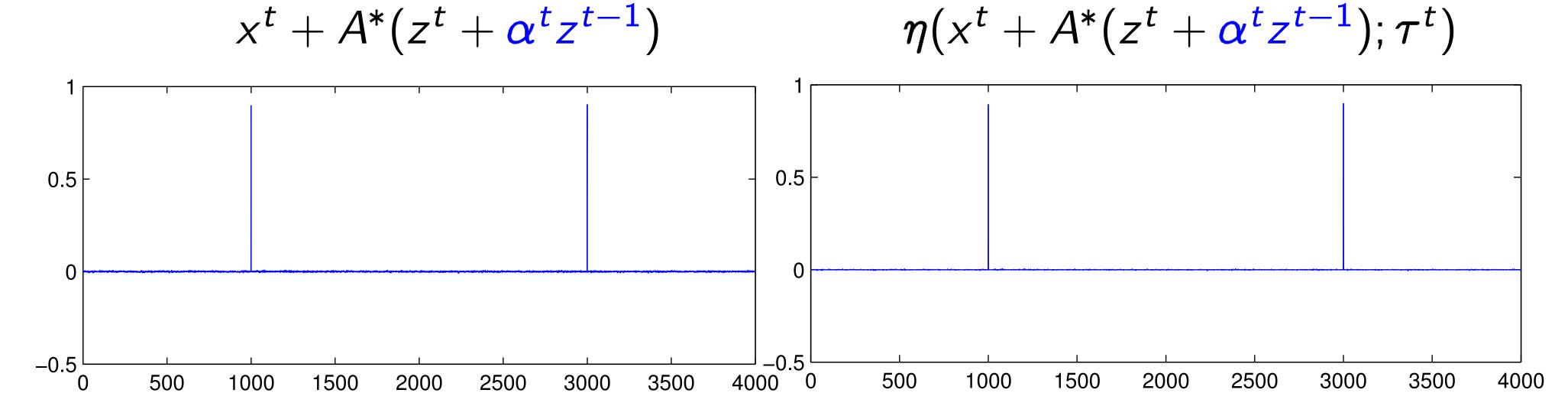


Iteration t = 3





AMP:



Weighted AMP

Given a support estimate $\tilde{T}\subseteq\{1,...,N\}$, assume $w_j=\omega<1$ for $j\in\tilde{T}$ and $w_j=1$ for $j\notin\tilde{T}$.

We incorporate this information into the AMP algorithm by the following weighted AMP algorithm:

$$x^{t+1} = \eta(x^{t} + A^{*}z^{t}; \hat{\tau}^{t}w),$$

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How does AMP work?

Assume $[w_1, ..., w_N]^T$ are the weights we use for the coefficients of the signal s.

Consider the following distribution over variables s_1, s_2, \ldots, s_N :

$$\mu(ds) = \frac{1}{Z} \prod_{i=1}^{N} \exp(-\beta \mathbf{w}_i | s_i|) \prod_{a=1}^{m} \delta_{\{y_a = (As)_a\}}, \tag{4}$$

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AMP and WAMP for seismic trace interpolation

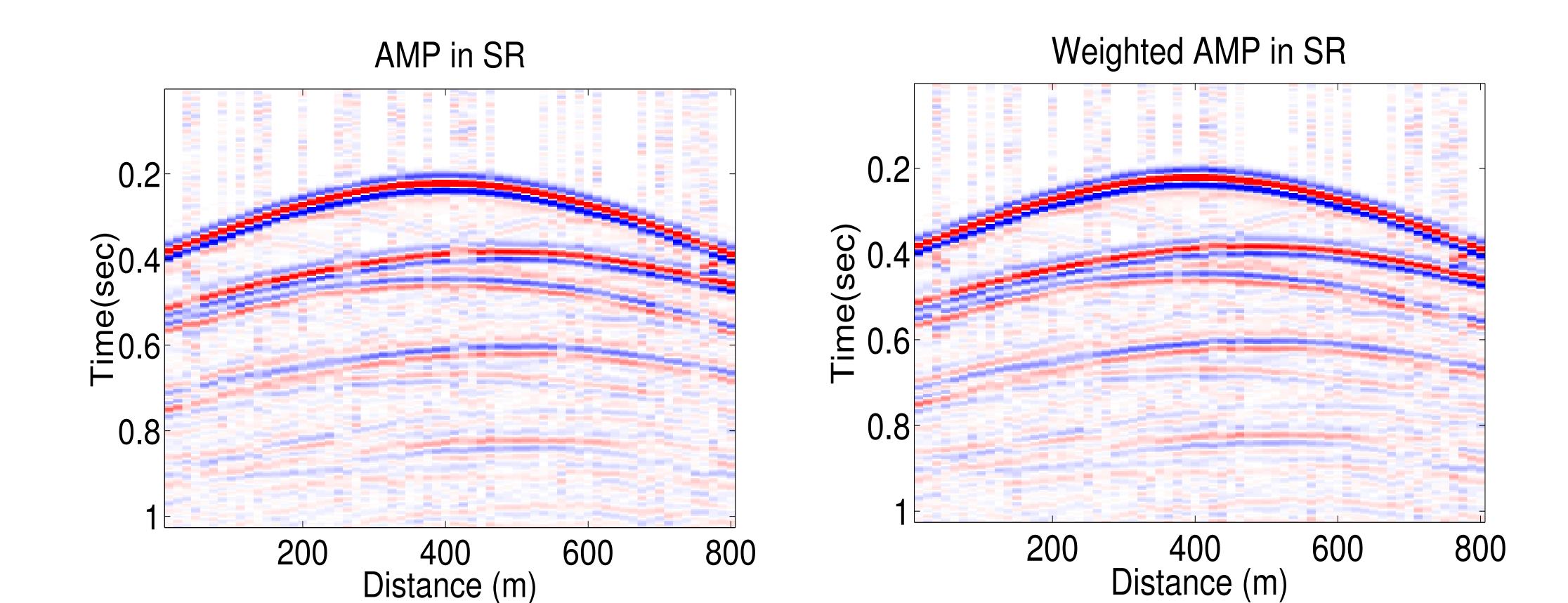
In order to use AMP and WAMP for seismic trace interpolation we use a 2-D DFT matrix as our sparsifying matrix.

Then $b = RMF_s^H F_s f$, where F_s is a 2-D DFT matrix in the source-receiver domain.

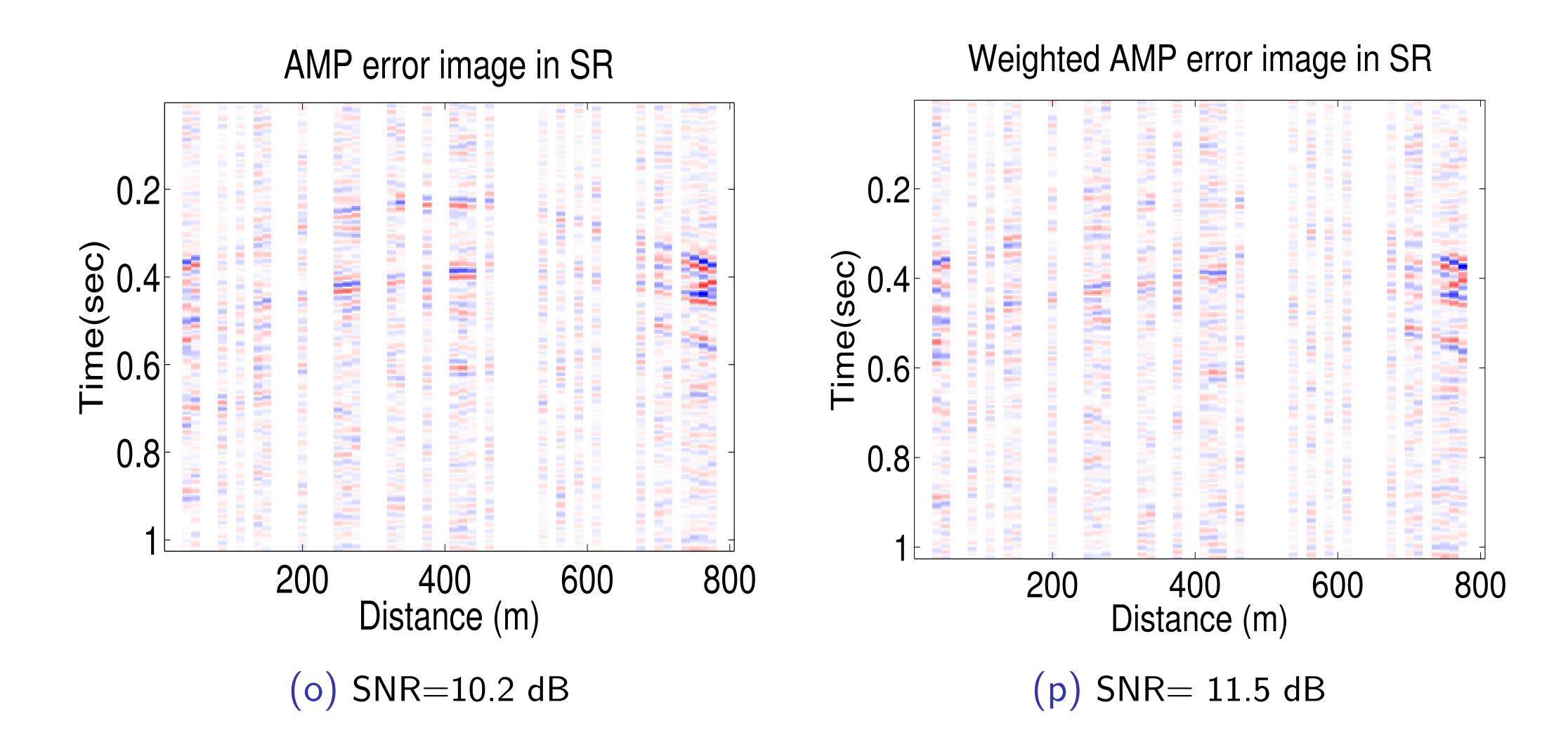
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Recovery error: AMP vs weighted AMP



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$$x^* \approx F_s f$$

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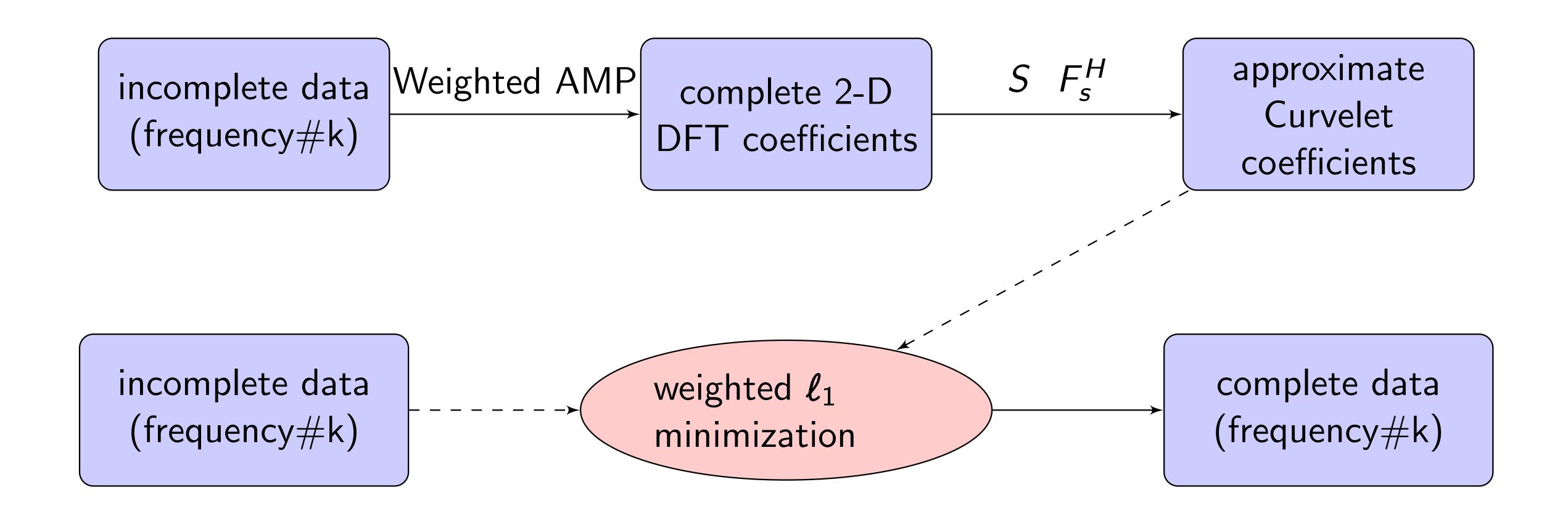
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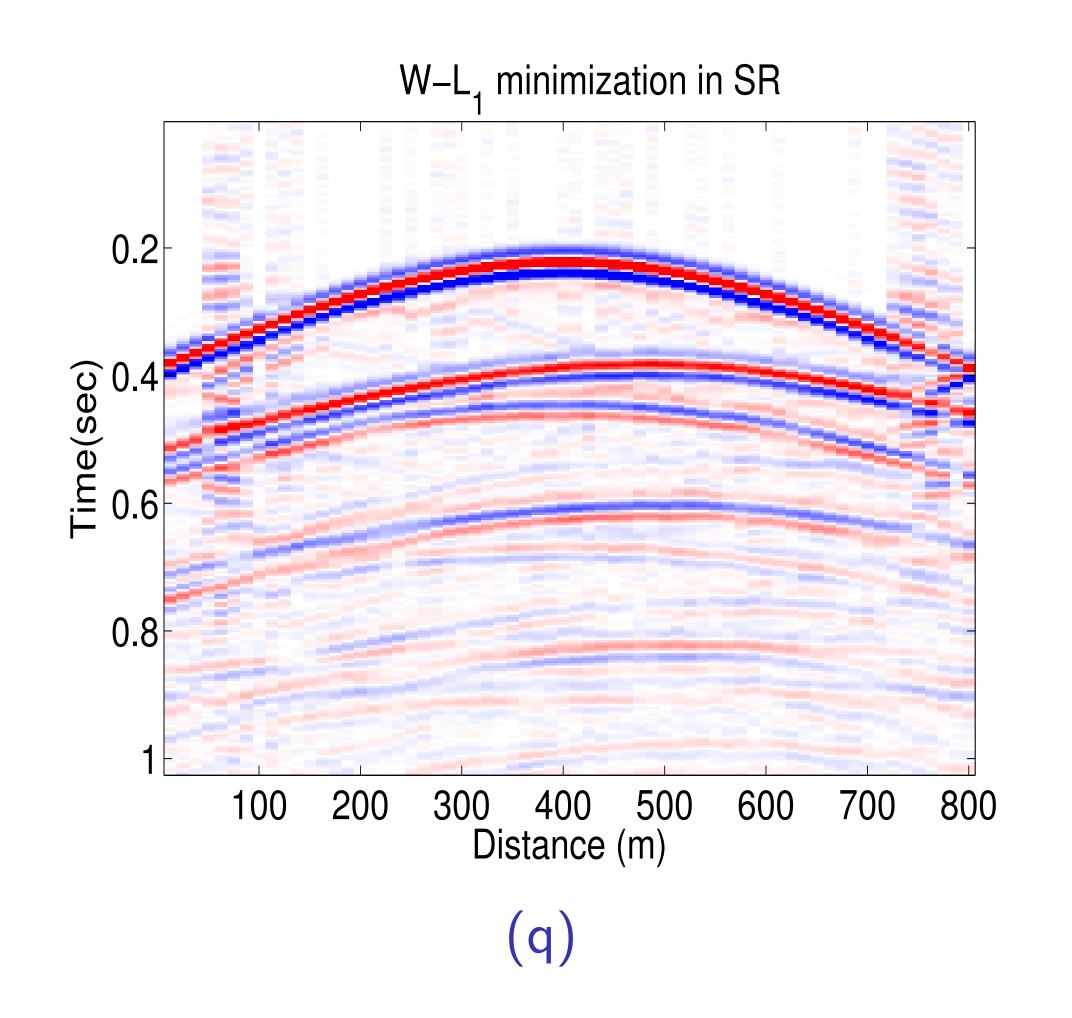
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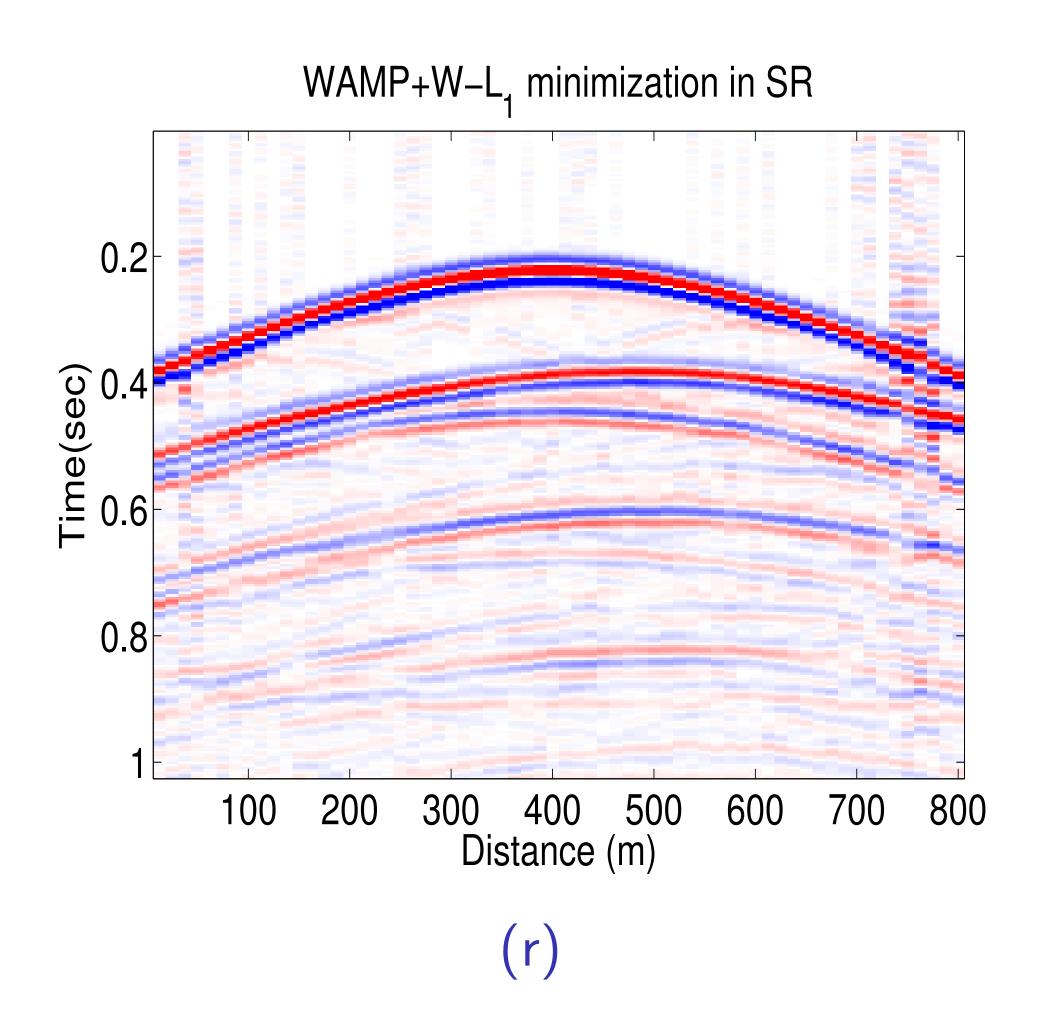
Flowchart of the 2-stage algorithm WAMP+weighted ℓ_1



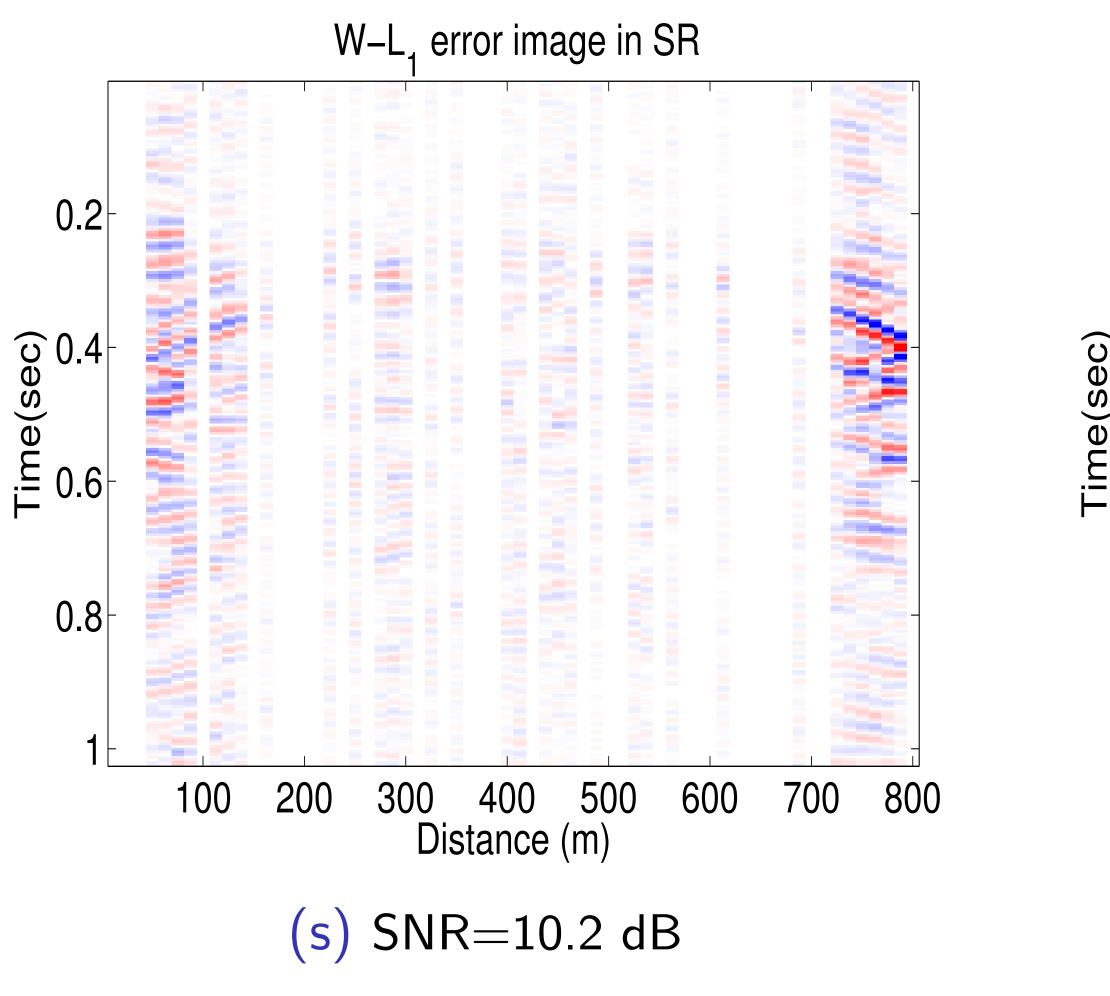
Weighted ℓ_1 vs 2-stage WAMP+weighted ℓ_1

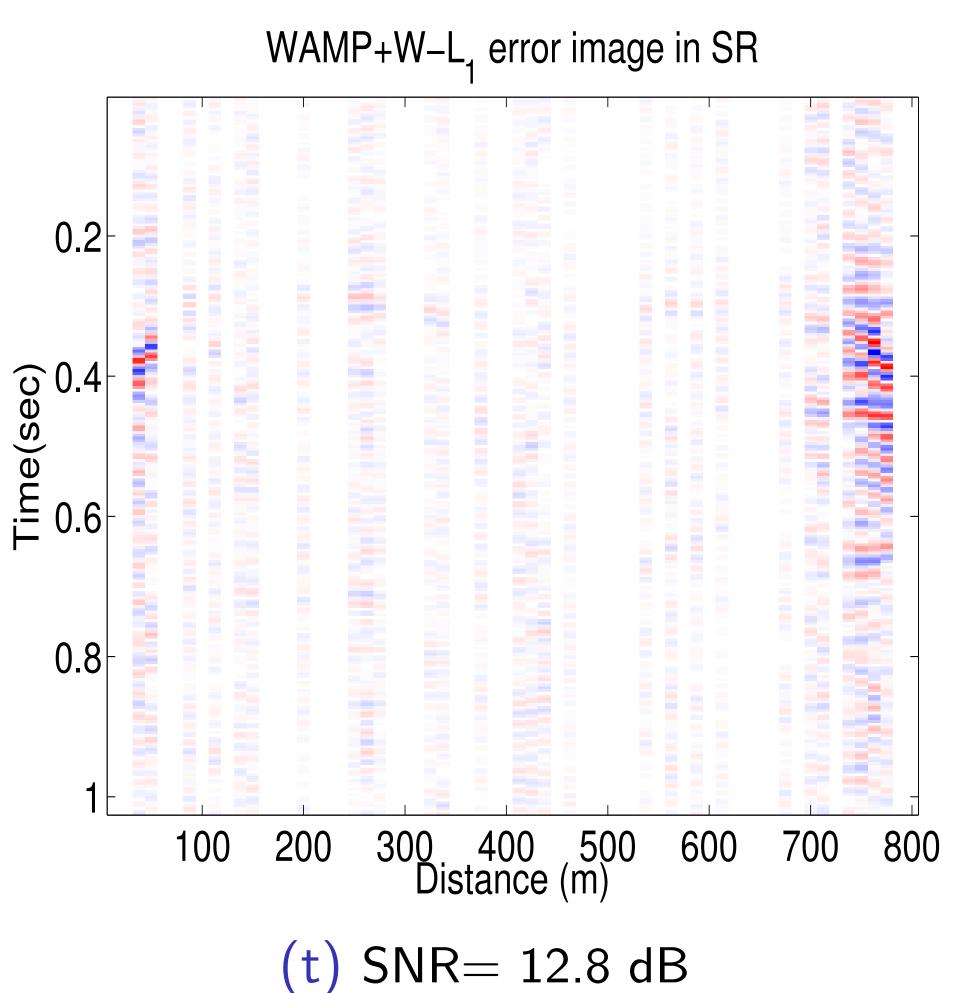
In the 2-stage algorithm, for each frequency slice we first apply a fast WAMP algorithm and use the result to derive new weights for weighted ℓ_1 minimizer with Curvelet coefficients.



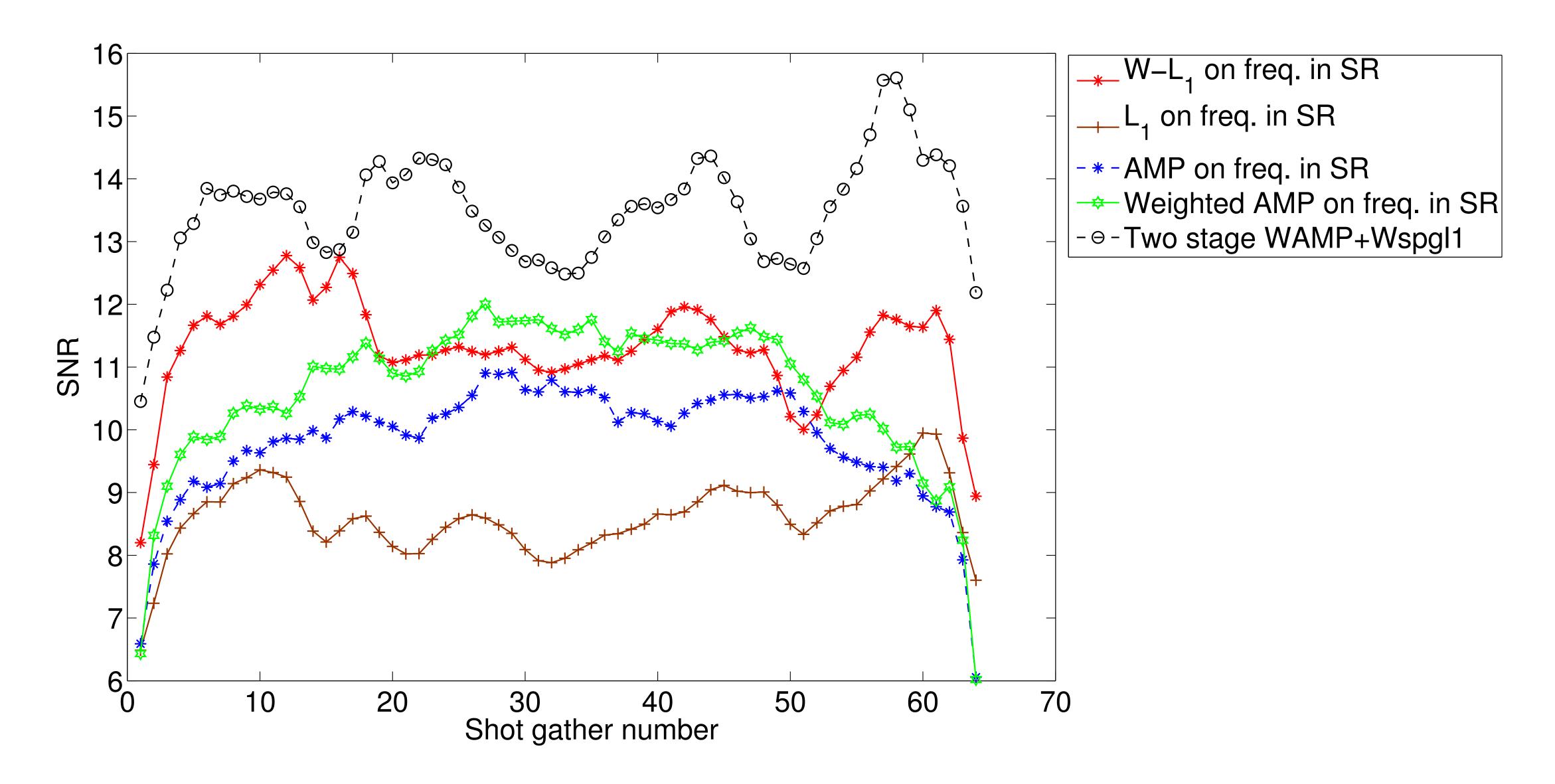


Recovery error: AMP vs weighted AMP





Comparison of shotgather SNRs



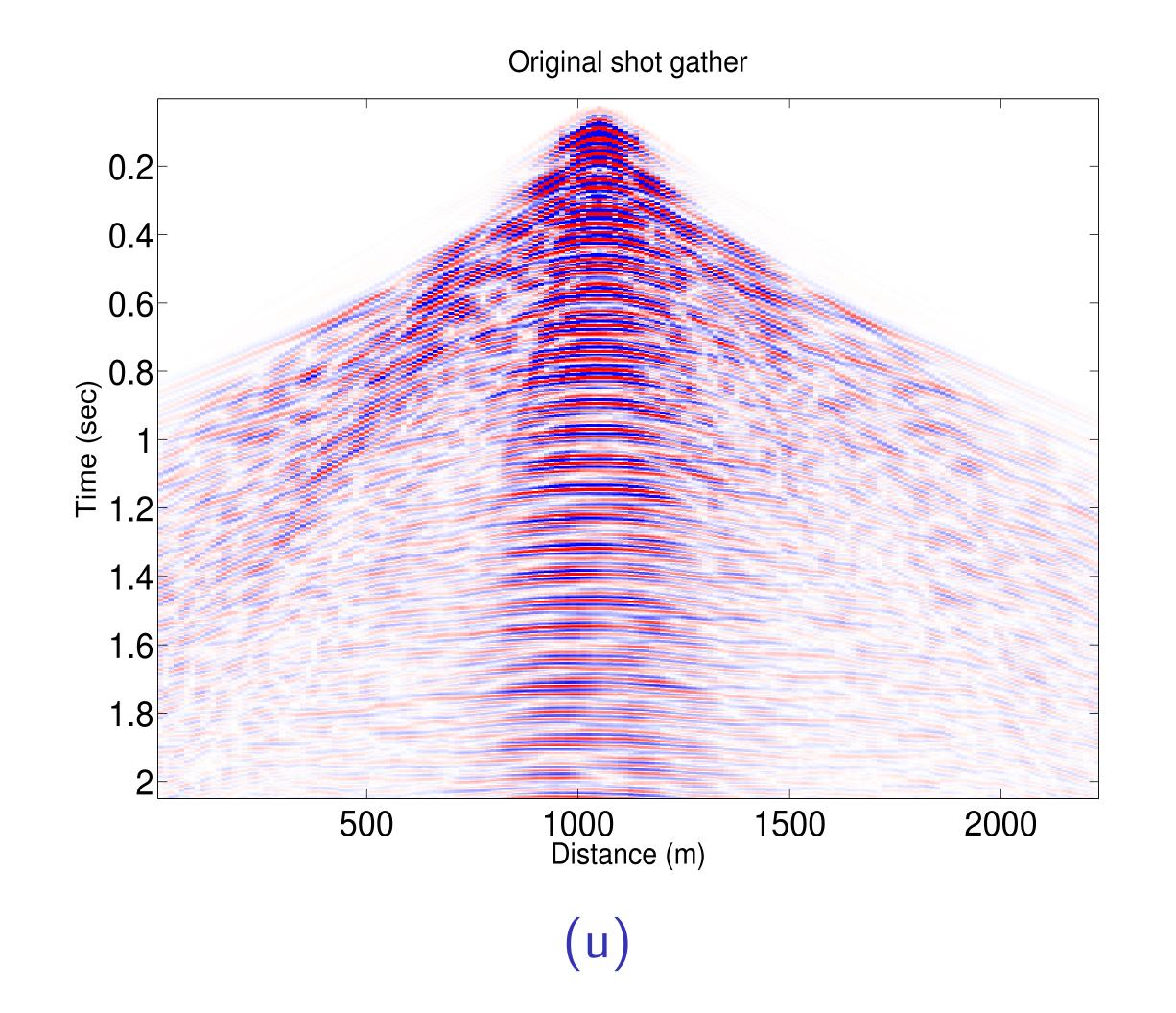
Results on the Gulf of Suez data

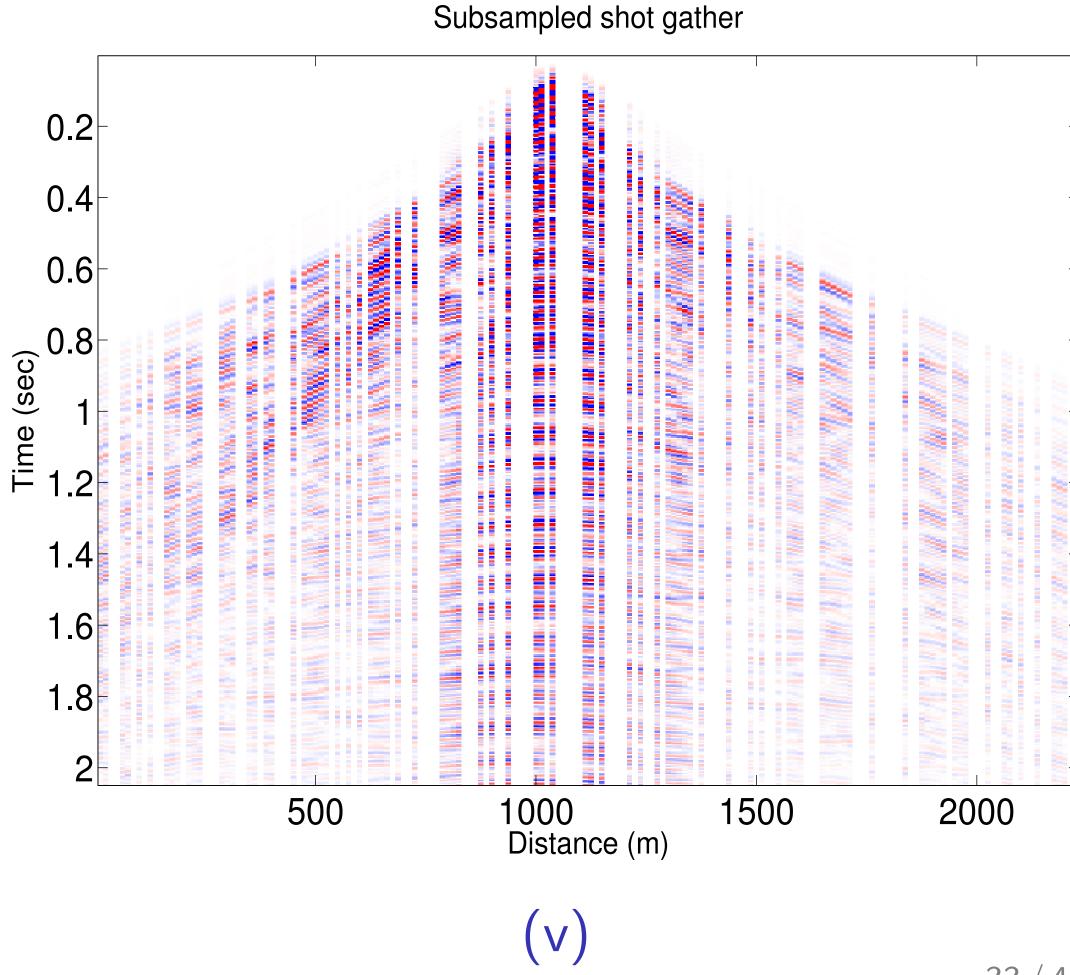
In the next slides we show the results of applying these algorithms on a seismic line from the Gulf of Suez.

The Seismic line at full resolution has $N_s = 178$ sources, $N_r = 178$ receivers with a sample distance of 12.5 meters, and $N_t = 512$ time samples acquired with a sampling interval of 4 milliseconds. Consequently, the seismic line contains samples collected in a 2s temporal window with a maximum frequency of 125 Hz.

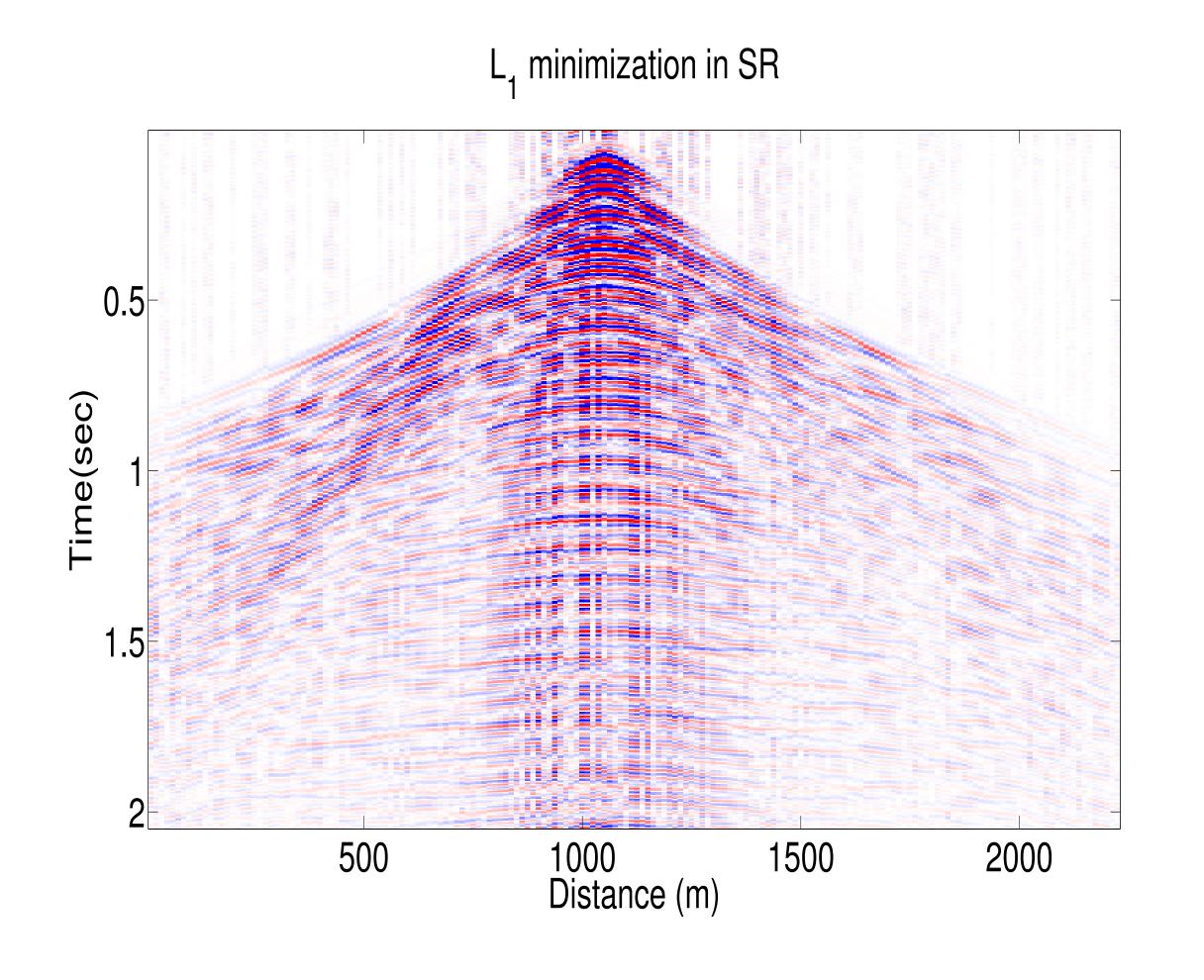
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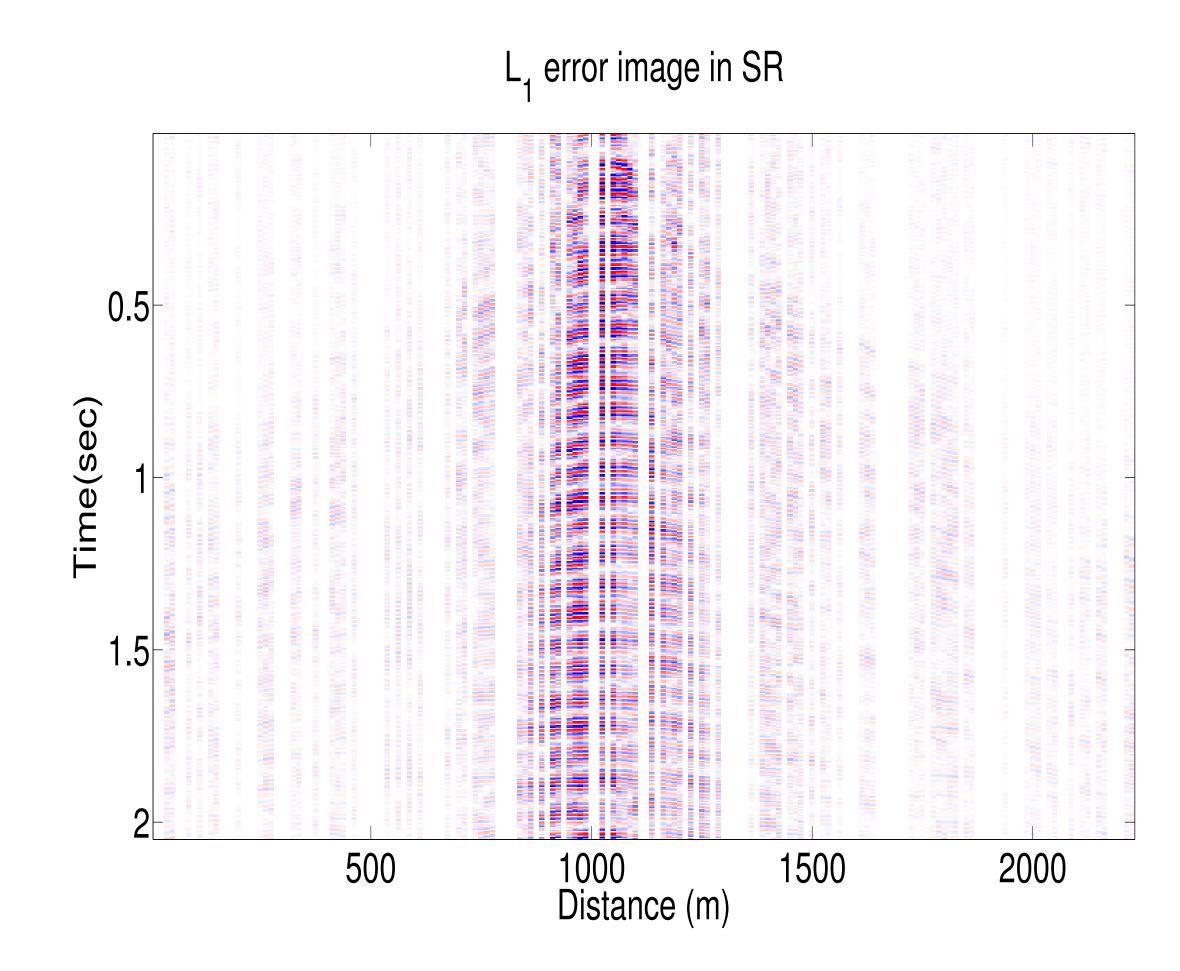
Shotgather number 84 from the seismic line:



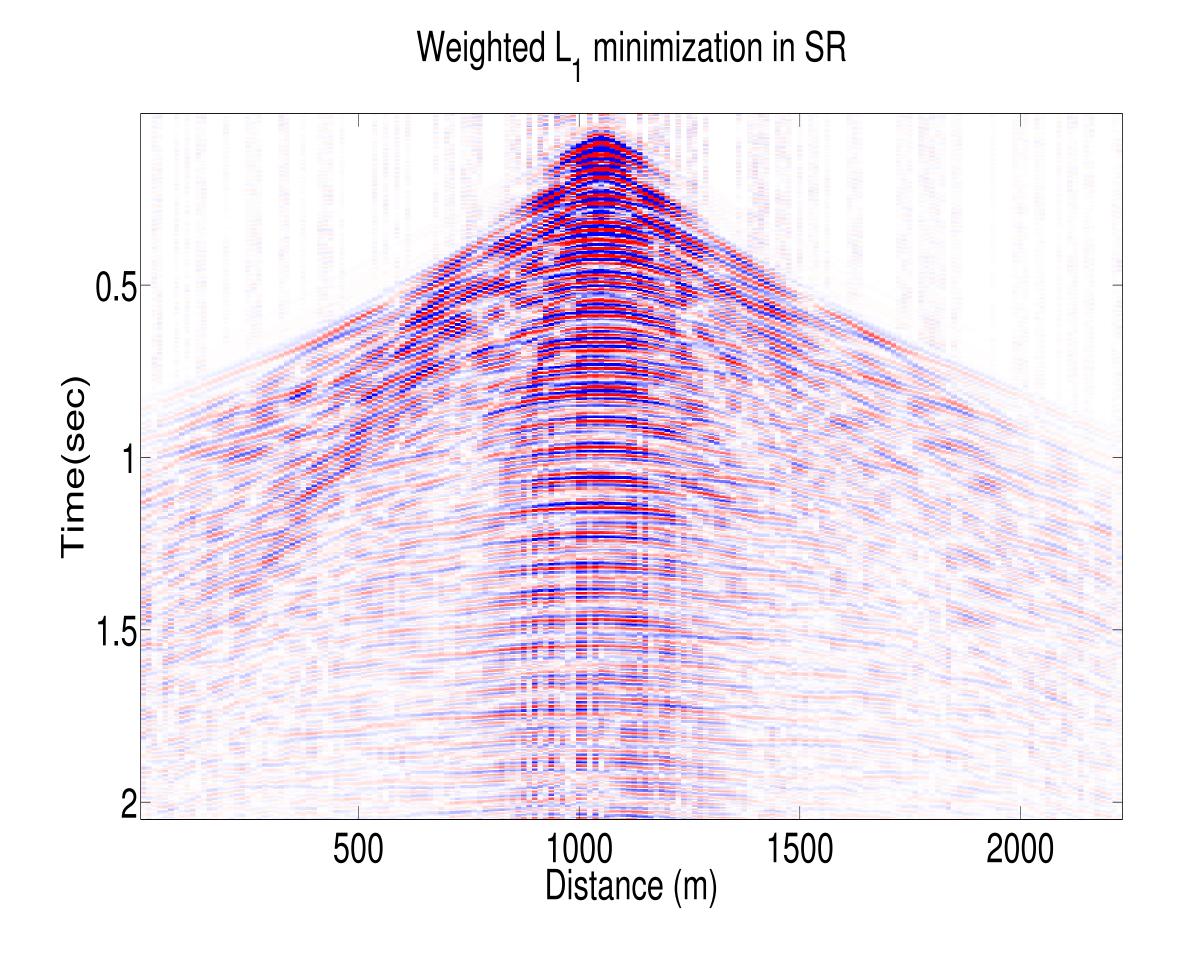


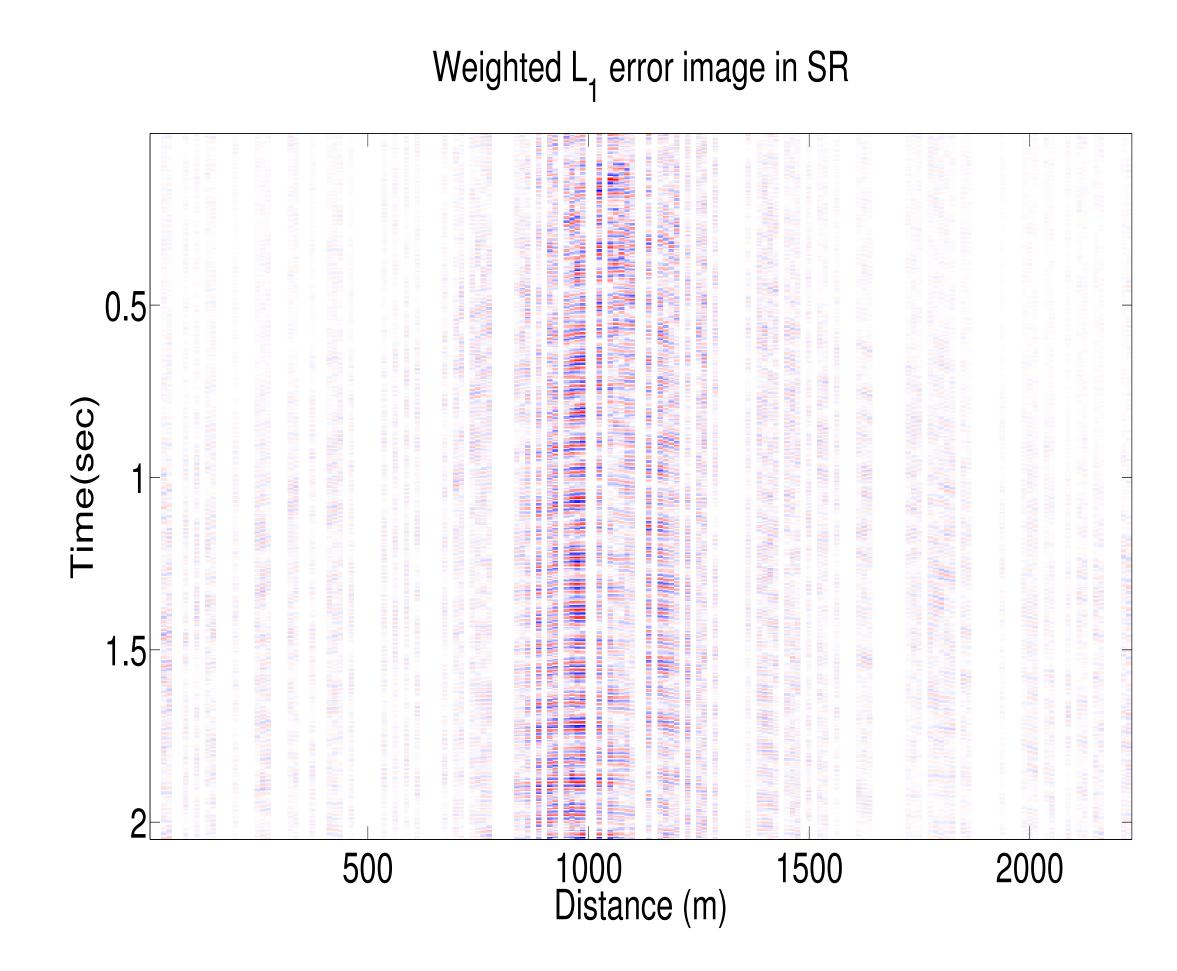
ℓ_1 minimization

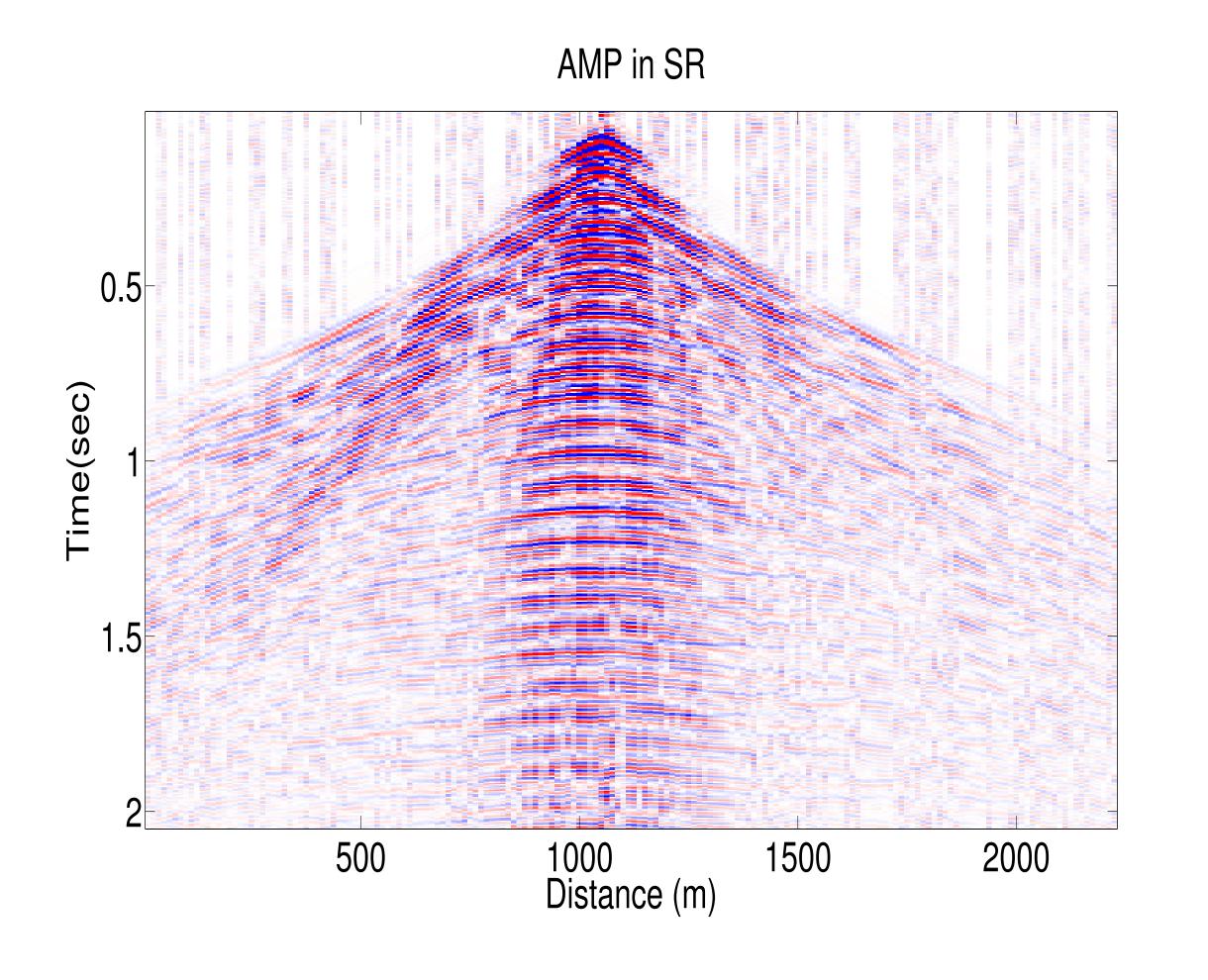


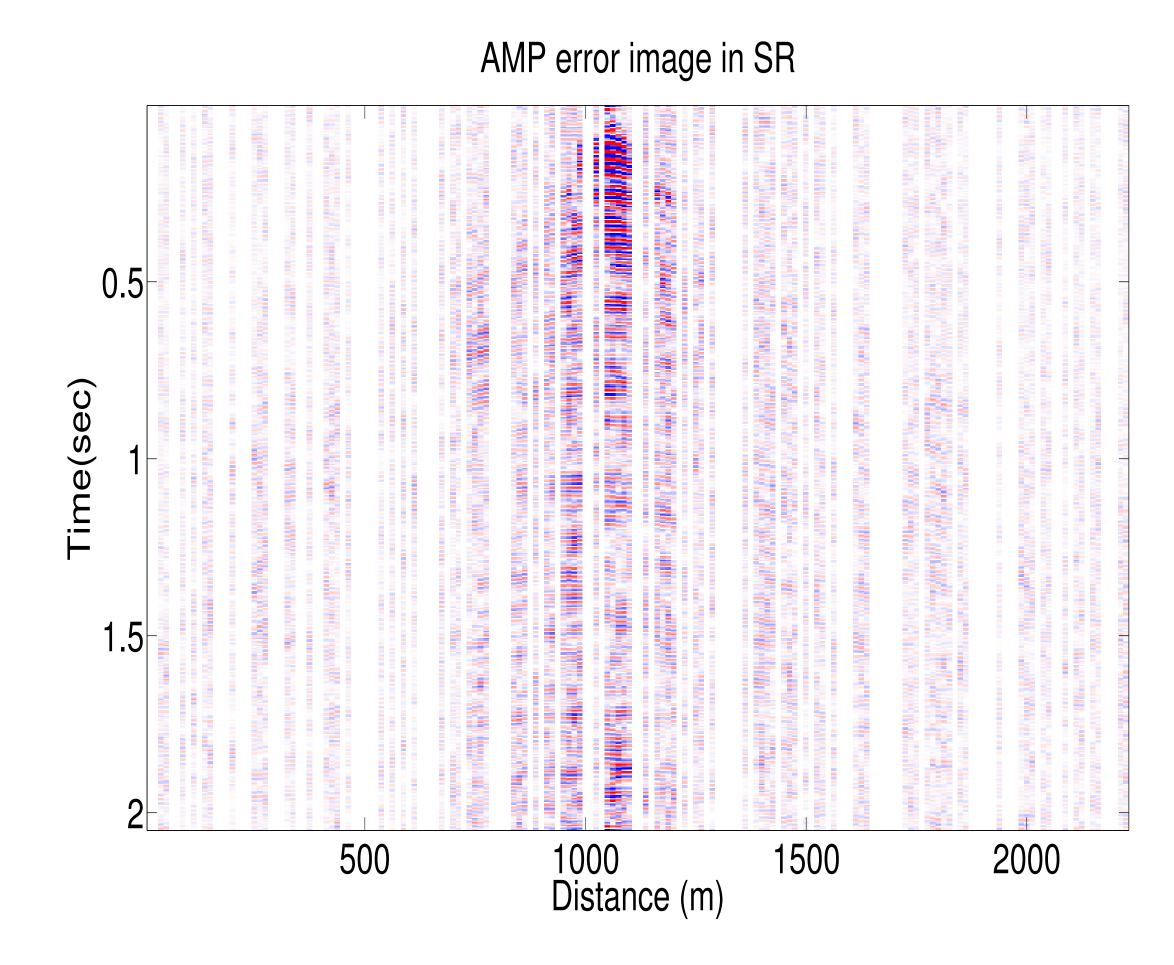


Weighted ℓ_1 minimization

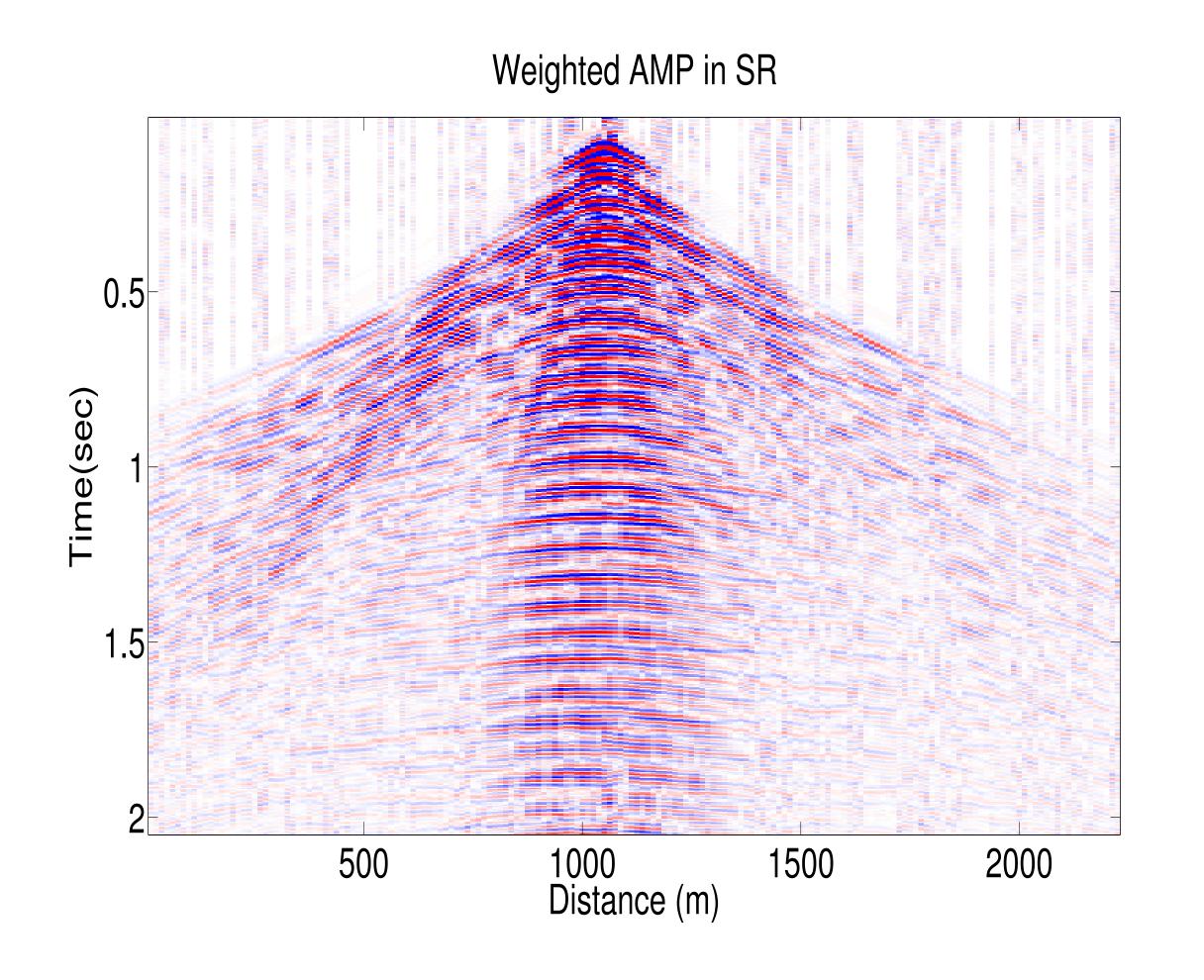


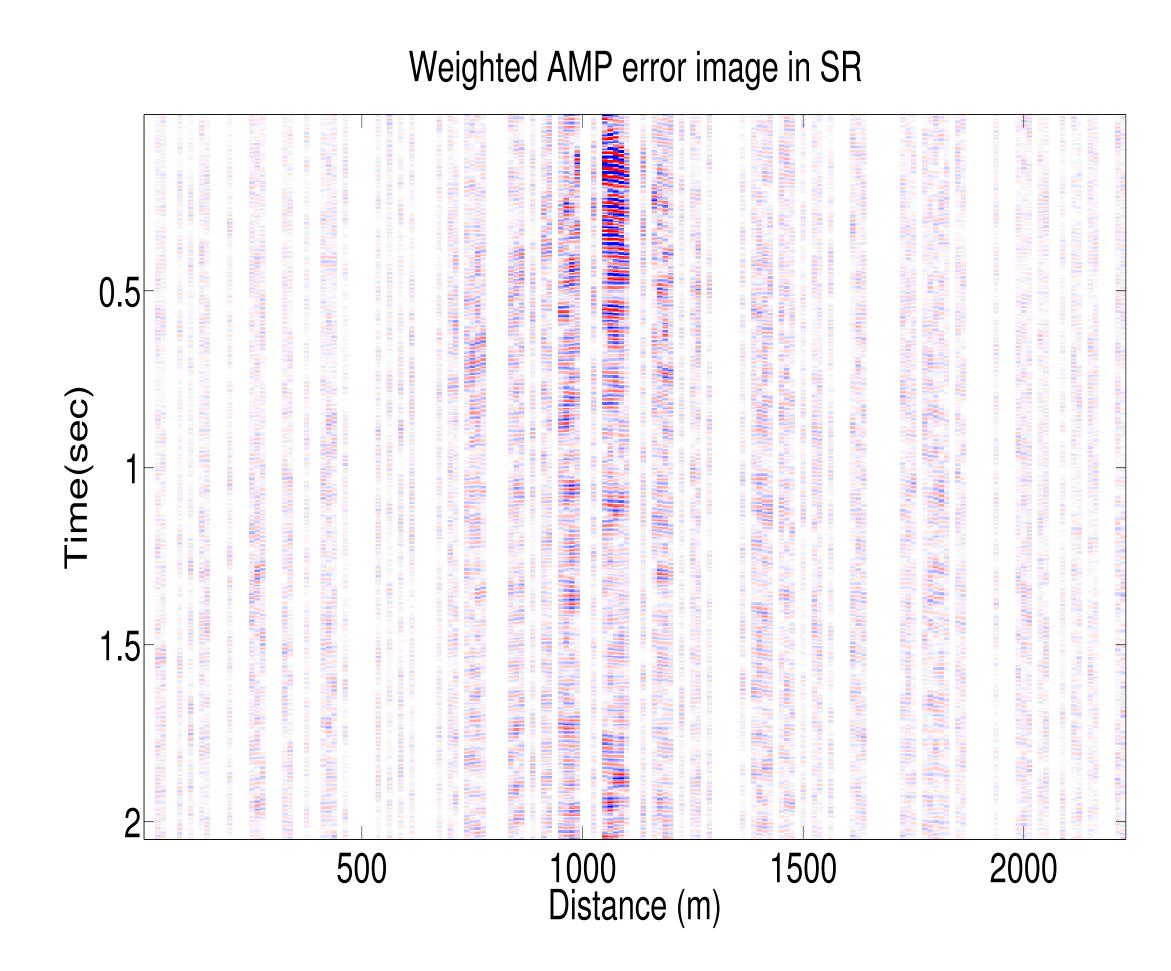




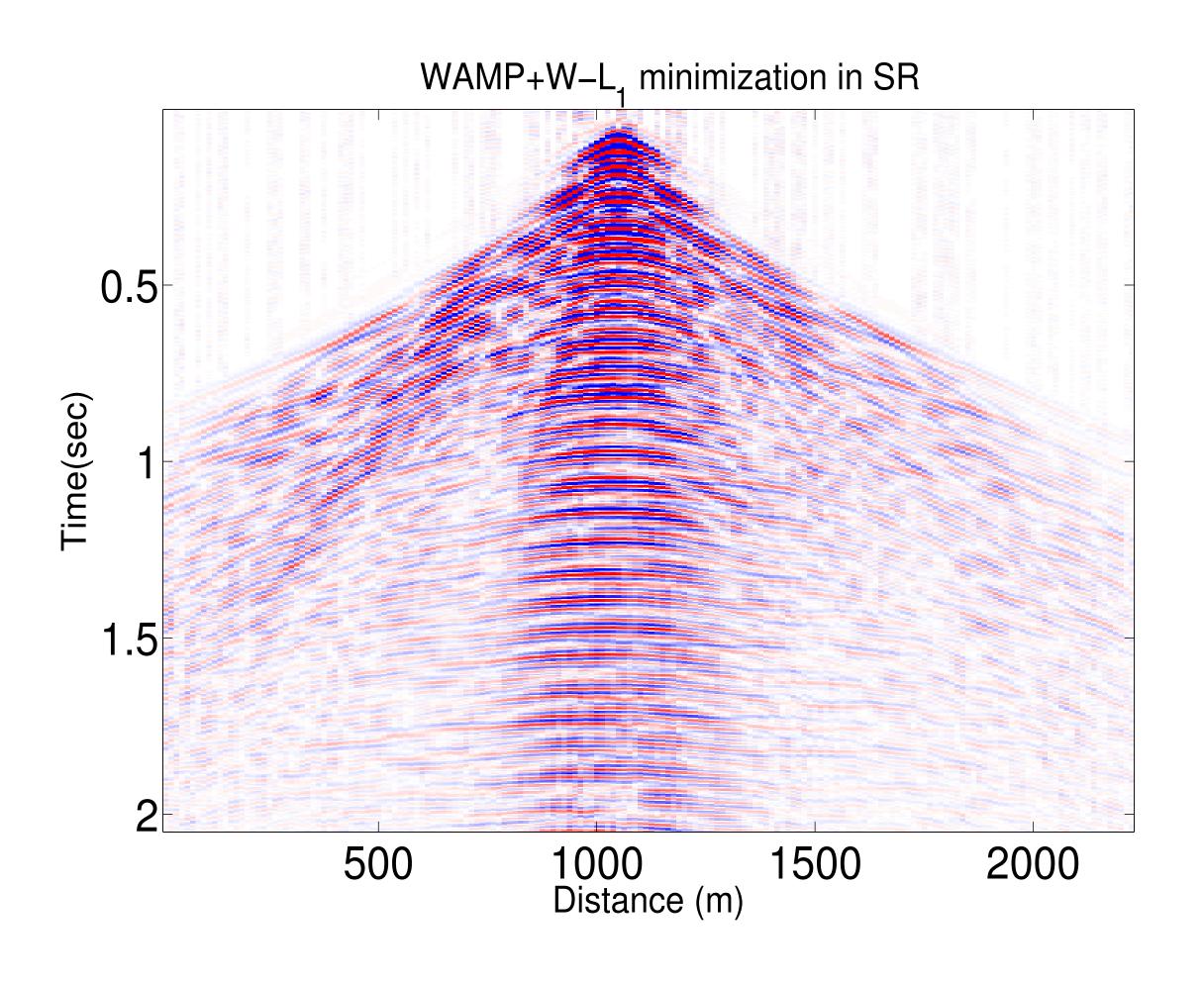


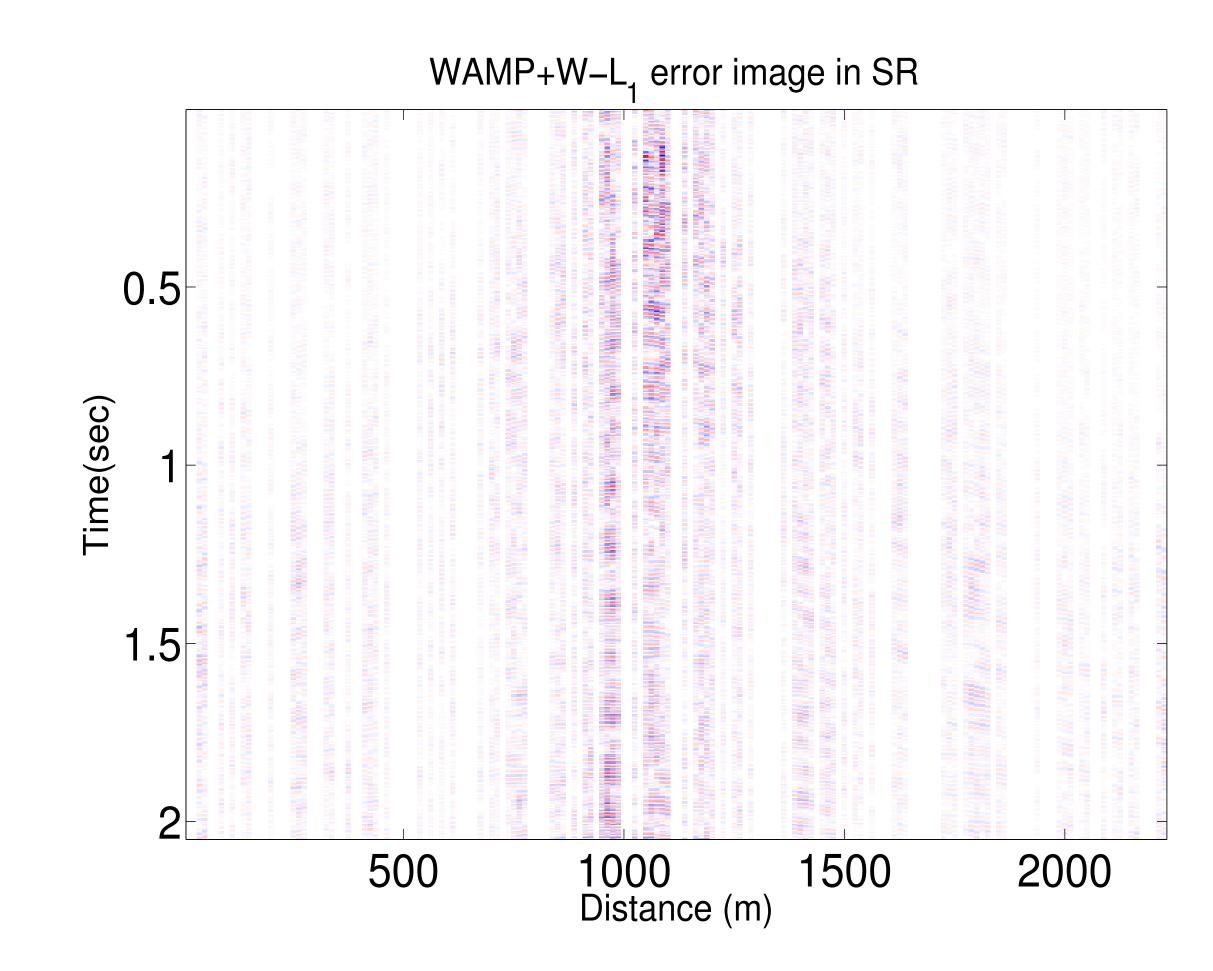
Weighted AMP



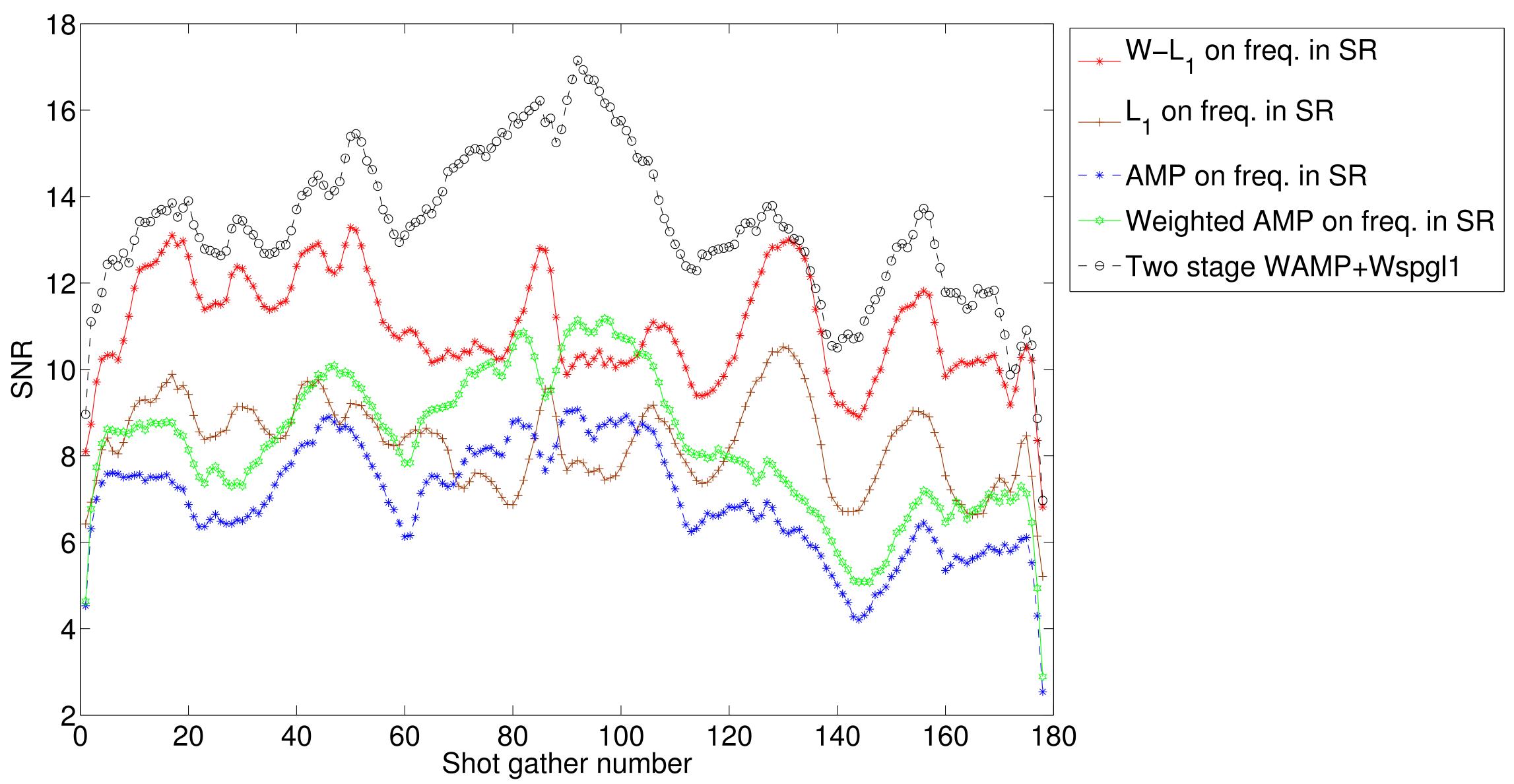


Weighted AMP+Weighted \(\ell_1 \)





Shotgather SNRs





Acknowledgements

Thank you for your attention!

https://www.slim.eos.ubc.ca/







This work was in part financially supported by the Natural Sciences and Engineering Research Council of Canada Discovery Grant (22R81254) and the Collaborative Research and Development Grant DNOISE II (375142-08). This research was carried out as part of the SINBAD II project with support from the following organizations: BG Group, BGP, BP, CGG, Chevron, ConocoPhillips, ION, Petrobras, PGS, Statoil, Total SA, WesternGeco, and Woodside.

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