# Robust 1-bit Compressive Sensing using Adaptive Outlier Pursuit

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Abstract-In compressive sensing (CS), the goal is to recover signals at reduced sample rate compared to the classic Shannon-Nyquist rate. However, the classic CS theory assumes the measurements to be real-valued and have infinite bit precision. The quantization of CS measurements has been studied recently and it has been shown that accurate and stable signal acquisition is possible even when each measurement is quantized to only one single bit. There are many algorithms proposed for 1bit compressive sensing and they work well when there is no noise in the measurements, e.g., there are no sign flips, while the performance is worsened when there are a lot of sign flips in the measurements. In this paper, we propose a robust method for recovering signals from 1-bit measurements using adaptive outlier pursuit. This method will detect the positions where sign flips happen and recover the signals using "correct" measurements. Numerical experiments show the accuracy of sign flips detection and high performance of signal recovery for our algorithms compared with other algorithms.

Index Terms—1-bit compressive sensing, adaptive outlier pursuit

### I. INTRODUCTION

THE theory of compressive sensing (CS) enables reconstruction of sparse or compressible signals from a small number of linear measurements relative to the dimension of the signal space [1], [2], [3], [4], [5]. In this setting, we have

$$y = \Phi x, \tag{1}$$

where  $x \in \mathbf{R}^N$  is the signal,  $\Phi \in \mathbf{R}^{M \times N}$  with M < Nis an underdetermined measurement system, and  $y \in \mathbf{R}^M$  is the set of linear measurements. It was demonstrated that Ksparse signals, i.e.,  $x \in \Sigma_K$  where  $\Sigma_K := \{x \in \mathbf{R}^N : ||x||_0 :=$  $|\operatorname{supp}(x)| \leq K\}$ , can be reconstructed exactly if  $\Phi$  satisfies the restricted isometry property (RIP) [6]. It was also shown that random matrices will satisfy the RIP with high probability if the entries are chosen according to independent and identically distributed (i.i.d.) Gaussian distribution.

Classic compressive sensing assumes that the measurements are real valued and have infinite bit precision. However, in practice, CS measurements must be quantized, i.e., each measurement has to be mapped from a real value (over a

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Furthermore, for some real world problems, severe quantization may be inherent or preferred. For example, in analogto-digital conversion (ADC), the acquisition of 1-bit measurements of an analog signal only requires a comparator to zero, which is an inexpensive and fast piece of hardware that is robust to amplification of the signal and other errors, as long as they preserve the signs of the measurements, see [13], [14]. In this paper, we will focus on the CS problem when 1-bit quantizer is used.

The 1-bit compressive sensing framework proposed in [13] is as follows. Measurements of a signal  $x \in \mathbf{R}^N$  are computed via

$$y = A(x) := \operatorname{sign}(\Phi x). \tag{2}$$

Therefore, the measurement operator  $A(\cdot)$  is a mapping from  $\mathbf{R}^N$  to the Boolean cube <sup>1</sup>  $\mathcal{B}^M := \{-1,1\}^M$ . We have to recover signal  $x \in \sum_K^* := \{x \in S^{N-1} : ||x||_0 \leq K\}$  where  $S^{N-1} := \{x \in \mathbf{R}^N : ||x||_2 = 1\}$  is the unit hyper-sphere of dimension N. Since the scale of the signal is lost during the quantization process, we can restrict the sparse signals to be on the unit hyper-sphere. Jacques et al. provided two flavors of results for the 1-bit CS framework [15]: 1) a lower bound is provided on the best achievable performance of this 1-bit CS framework, and if the elements of  $\Phi$  are drawn randomly from i.i.d. Gaussian distribution or its rows are drawn uniformly from the unit sphere, then the solution will have bounded error on the order of the optimal lower bound. 2) A condition on the mapping A, binary  $\epsilon$ -stable embedding (B $\epsilon$ SE), that enables stable reconstruction is given to characterize the reconstruction performance even when some of the measurement signs have changed (e.g., due to noise in the measurements).

Since this problem was introduced and studied by Boufounos and Baraniuk in 2008 [13], it has been studied by many people and several algorithms have been developed [13], [15], [16], [17], [18], [19], [20]. Binary iterative hard thresholding (BIHT) [15] is shown to perform better than other algorithms such as matching sign pursuit (MSP) [16] and restricted-step shrinkage (RSS) [18] in both reconstruction error as well as consistency, see [15] for more details. The experiment in [15] shows that the one-sided  $\ell_1$  objective

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<sup>&</sup>lt;sup>1</sup>Generally, the *M*-dimensional Boolean cube is defined as  $\{0, 1\}^M$ . Without loss of generality, we use  $\{-1, 1\}^M$  instead.

(BIHT) performs better when there are only a few errors, and the one-sided  $\ell_2$  objective (BIHT- $\ell_2$ ) performs better when there are significantly more errors, which implies that BIHT- $\ell_2$  is useful when the measurements contain significant noise that might cause a large number of sign flips.

In practice, there will always be noise in the measurements during acquisition and transmission, therefore, a robust algorithm for 1-bit compressive sensing when the measurements flip their signs is strongly needed. One possible way to build this robust algorithm is to introduce an outlier detection technique.

There are many applications where the accurate detection of outliers is needed. For example, when an image is corrupted by random-valued impulse noise, the corrupted pixels are useless in image denoising. There are some methods (e.g., adaptive center-weighted median filter (ACWMF) [21]) for detecting the damaged pixels. But these methods will miss quit a lot of real noise and false-hit some noise-free pixels when the noise level is high. In [22], we proposed a method to adaptively detect the noise pixels and restore the image with  $\ell_0$  minimization. Instead of detecting the damaged pixels before recovering the image, we iteratively restore the image and detect the damaged pixels. This idea works really well for impulse noise removal. In this 1-bit compressive sensing framework, when there is a sign flip in one measurement, this measurement will worsen the reconstruction performance. If we can detect all the measurements with sign flips, then we can change the signs for these measurements and improve the reconstruction performance a lot. However, it is much more difficult than detecting impulse noise and there is no method for detecting sign flips, but we can still utilize the idea in [22] to adaptively find the sign flips. In this paper, we will introduce a method for robust 1-bit compressive sensing which can detect the sign flips and reconstruct the signals with very high accuracy even when there are a large number of sign flips.

This paper is organized as follows. We will introduce several algorithms for reconstructing the signal and detecting the sign flips in section II. Section III studies the case when the noise information is not given. The performance of these algorithms is illustrated in section IV with comparison to BIHT and BIHT- $\ell_2$ . We will end this work by a short conclusion.

## II. ROBUST 1-BIT COMPRESSIVE SENSING USING Adaptive Outlier Pursuit

Binary iterative hard thresholding (BIHT or BIHT- $\ell_2$ ) in [15] is the algorithm for solving

$$\min_{x} \quad \sum_{i=1}^{M} \phi(y_i, (\Phi x)_i) \\
\text{subject to:} \quad \|x\|_2 = 1, \quad \|x\|_0 \le K,$$
(3)

where  $\phi$  is the one-sided  $\ell_1$  (or  $\ell_2$ ) objective:

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$$\phi(x,y) = \begin{cases} 0, & \text{if } x \cdot y > 0, \\ |x \cdot y| \text{ (or } |x \cdot y|^2/2), & \text{otherwise.} \end{cases}$$
(4)

The high performance of BIHT is demonstrated when all the measurements are noise-free. However when there are a lot of sign flips, the performance of BIHT and BIHT- $\ell_2$  is worsened

by the noisy measurements. There is no method to detect the sign flips in the measurements, but adaptively finding the sign flips and reconstructing the signals can be combined together as in [22] to obtain better performance.

Let us assume firstly that the noise level (the ratio of the number of sign flips over the number of measurements for 1-bit compressive sensing) is provided. Based on this information, we can choose a proper integer L such that at most L elements of the total measurements are wrongly detected (having sign flips). For measurements  $y \in \{-1, 1\}^M$ ,  $\Lambda \in \mathbf{R}^M$  is a binary vector denoting the "correct" data:

$$\Lambda_i = \begin{cases} 1, & \text{if } y_i \text{ is "correct",} \\ 0, & \text{otherwise.} \end{cases}$$
(5)

According to the assumption, we have  $\sum_{i=1}^{M} (1 - \Lambda_i) \leq L$ .

Introducing  $\Lambda$  into the old problem solved by BIHT, we have the following new problem with unknown variable x and  $\Lambda$ :

$$\min_{x,\Lambda} \sum_{i=1}^{M} \Lambda_i \phi(y_i, (\Phi x)_i) \\
\text{s.t.} \sum_{i=1}^{M} (1 - \Lambda_i) \le L, \\
\Lambda_i \in \{0, 1\} \quad i = 1, 2, \cdots, M, \\
\|x\|_2 = 1, \quad \|x\|_0 \le K.$$
(6)

The above model can also be interpreted in the following way. Let us consider the noisy measurement y as the sign of  $\Phi x$  with additive unknown noise n, i.e.  $y = \text{sign}(\Phi x + n)$ . Though the binary measurement is robust to noise as long as the sign does not change, there exist some  $n_i$ 's such that the corresponding measurements change. In our problem, only a few measurements are corrupted, and only these corresponding  $n_i$ 's are important. Therefore, n can be considered as sparse noise with nonzero entries at these locations, and we have to recover the signal x from sparse corrupted measurements [23], [24], even when the measurements is acquired by taking the sign of  $\Phi x + n$ . This equivalent problem is

$$\min_{\substack{x,n \\ s.t.}} \sum_{i=1}^{M} \phi(y_i, (\Phi x)_i + n_i) \\
\text{s.t.} \quad \|n\|_0 \le L, \\
\|x\|_2 = 1, \quad \|x\|_0 \le K.$$
(7)

The equivalence is described in the appendix.

The problem defined in (6) is non-convex and has both continuous and discrete variables. It is difficult to solve it together, thus we use alternating minimization method, which separates the energy minimization over x and  $\Lambda$  into two steps:

• Fix  $\Lambda$  and solve for x:

$$\min_{x} \quad \sum_{i=1}^{M} \Lambda_{i} \phi(y_{i}, (\Phi x)_{i}) \\
\text{s.t.} \quad \|x\|_{2} = 1, \quad \|x\|_{0} \le K.$$
(8)

This is the same as (3) with revised  $\Phi$  and y. We only need to use the *i*th rows of  $\Phi$  and y where  $\Lambda_i = 1$ .

• Fix x and update  $\Lambda$ :

$$\min_{\Lambda} \sum_{\substack{i=1\\M}}^{M} \Lambda_i \phi(y_i, (\Phi x)_i) \\
\text{s.t.} \sum_{\substack{i=1\\\Lambda_i \in \{0, 1\}}}^{M} (1 - \Lambda_i) \leq L, \\
\Lambda_i \in \{0, 1\} \quad i = 1, 2, \cdots, M.$$
(9)

This problem is to choose M - L elements with least sum from M elements  $\{\phi(y_i, (\Phi x)_i)\}_{i=1}^M$ . Given an xestimated from (8), we can update  $\Lambda$  in one step:

$$\Lambda_i = \begin{cases} 0, & \text{if } \phi(y_i, (\Phi x)_i) \ge \tau, \\ 1, & \text{otherwise.} \end{cases}$$
(10)

where  $\tau$  is the  $L^{th}$  largest term of  $\{\phi(y_i, (\Phi x)_i)\}_{i=1}^M$ . If the  $L^{th}$  and  $(L+1)^{th}$  larges terms are equal, then we can choose any  $\Lambda$  such that  $\sum_{i=1}^M \Lambda_i = M - L$  and

$$\min_{i,\Lambda_i=0}\phi(y_i,(\Phi x)_i) \ge \max_{i,\Lambda_i=1}\phi(y_i,(\Phi x)_i).$$

Since for each step, the updated  $\Lambda$  identifies the outliers, this method is named as adaptive outlier pursuit (AOP). When L = 0, this is exactly the BIHT proposed in [15]. Our algorithm is as follows:

# Algorithm 1 AOP

**Input:**  $\Phi \in \mathbf{R}^{M \times N}$ ,  $y \in \{-1, 1\}^M$ , K > 0,  $L \ge 0$ ,  $\alpha > 0$ , Miter > 0Initialization:  $x^0 = \Phi^T y / \| \Phi^T y \|, k = 0, \Lambda = \mathbf{1} \in \mathbf{R}^M$ , Loc = 1 : M, tol = inf, TOL = inf. while  $k \leq$  Miter and  $L \leq$  tol **do** Compute  $\beta^{k+1}$  $= x^k + \alpha \Phi(\text{Loc},:)^T(y(\text{Loc}) \operatorname{sign}(\Phi(\operatorname{Loc},:)x^k)).$ Update  $x^{k+1} = \eta_K(\beta^{k+1}),$ Set tol =  $||y - A(x^{k+1})||_0$ . if tol  $\leq$ TOL then, Compute  $\Lambda$  with (10). Update Loc to be the location of 1-entries of  $\Lambda$ . Set TOL = tol. end if k = k + 1.end while return  $x^k / ||x^k||$ .

 $\eta_K(v)$  computes the best *K*-term approximation of *v* by thresholding. Since  $y_i \in \{-1, 1\}$ , once we find the locations of the errors, instead of deleting these data, we can also "correct" them by flipping their signs. Hence *x* can also be updated with  $\Phi$  and this new measurements. This algorithm with changing signs is called AOP with flips.

**Remark**: Similar to BIHT- $\ell_2$ , we can also choose the onesided  $\ell_2$  objective instead of  $\ell_1$  objective and obtain two other algorithms.

## III. THE CASE WITH L UNKNOWN

In previous section, we assume that L, the number of corrupted measurements, is known in advance. However in real world applications there are cases when no pre-knowledge

about the noise is given. If L is chosen smaller or larger than the true value, the performance of these algorithms will get worse. As shown numerically in section IV, when L is less than the true value, even if the L detections are completely correct, some sign flips still stay in the measurements. On the other hand, some correct measurements will be lost if L is too large, and the problem will have more solutions if the number of total measurements is not large enough, which will affect the accuracy of the algorithm. Therefore, in this scenario we have to apply an L detection skill to find an L which is not far from the true value.

When no noise information is given, the following procedure can be applied to predict L. The first-phase preparation is to do extensive experiments on simulated data with known L and record the Hamming distances between A(x) and noisy y of BIHT- $\ell_2$  and AOP. Here we can simply use the results in our first experiment in section IV. The average of the results describes nicely the behavior of these two algorithms at different noise levels. Hence a formula can be derived to predict the Hamming distance of AOP based on the results obtained by BIHT- $\ell_2$ . This could be a fair initial guess for the noise level, and we can derive an L based on the result, labeled as  $L_0$ . Then we calculate  $L_t = ||A(x) - y||_0$  using the result x gained by AOP with  $L_0$  as the input for L. If  $L_t$ is greater than  $L_0$ , which means that  $L_0$  is too small while  $L_t$  is too large, we set  $L_t$  as the upper bound  $L_{\max}$  and  $L_0$ as the lower bound  $L_{\min}$ . Otherwise, if  $L_t$  is smaller than or equal to  $L_0$ , which means  $L_0$  may be too large, we use  $\mu L_0$  $(0 < \mu < 1)$  as the new  $L_0$  to look for new  $L_t$ . We will keep doing this until  $L_t$  is greater than  $L_0$ . Then the previous  $L_0$  is defined as the upper bound  $L_{\text{max}}$  and the new  $L_0$  is defined as the lower bound  $L_{\min}$ . This is just one method for finding lower and upper bounds for L, and there are certainly other possible ways to decide the bounds. Then we use the bisection method to find a better L. The mean of  $L_{\text{max}}$  and  $L_{\min}$  ( $L_{mean}$ ) is then used as input to derive  $L_t$  with AOP. If  $L_t$  is greater than  $L_{mean}$ , we update  $L_{min}$  with  $L_{mean}$ . Otherwise,  $L_{mean}$  is set as  $L_{max}$ . This bisection method is applied to update these two bounds until  $L_{\rm max} - L_{\rm min} \leq 1$ . The final  $L_{\min}$  is our input L.

### **IV. NUMERICAL RESULTS**

In this section we use several numerical experiments to demonstrate the effectiveness of AOP algorithms. Here AOP is implemented in the following four ways: 1) AOP with one-sided  $\ell_1$  objective (AOP); 2) AOP with flips and one-sided  $\ell_1$  objective (AOP-f); 3) AOP with one-sided  $\ell_2$  objective (AOP- $\ell_2$ ); 4) AOP with flips and one-sided  $\ell_2$  objective (AOP- $\ell_2$ -f). The four algorithms, together with BIHT and BIHT- $\ell_2$ , are studied and compared in the following experiments.

The setup for our experiments is as follows. We first generate a matrix  $\Phi \in \mathbf{R}^{M \times N}$  whose elements follow i.i.d. Gaussian distribution. Then we generate the original K-sparse signal  $x^* \in \mathbf{R}^N$ . Its non-zero entries are drawn from standard Gaussian distribution and then normalized to have norm 1.  $y^* \in \{-1, 1\}^M$  is computed by  $A(x^*)$ .



(c) Hamming error between A(x) and(d) Hamming distance between A(x)  $A(x^*)$  and y

Fig. 1: Algorithm comparison on corrupted data with different noise levels. (a) average SNR vs noise level, (b) average angular error vs noise level, (c) average Hamming error between A(x) and  $A(x^*)$  vs noise level, (d) average Hamming distance between A(x) and noisy measurements y vs noise level. AOP proves to be more robust to measurement sign flips compared with BIHT.

## A. noise levels test

In our first experiment, we set M = N = 1000, K = 10, and examine the performance of these algorithms on data with different noise levels. Here in each test, we choose a few measurements at random and flip their signs. The noise level is between 0% and 10% and we assume it is known in advance. For each level, we perform 100 trials and record the average signal-to-noise ratio (SNR), average reconstruction angular error for each reconstructed signal x with respect to  $x^*$ , average Hamming error between A(x) and  $A(x^*)$  and average Hamming distance between A(x) and the noisy measurements y. Here SNR is denoted by  $10 \log_{10}(||x||^2/||x-x^*||^2)$ , angular error is defined as  $\arccos\langle x, x^* \rangle / \pi$ , Hamming error stands for  $||A(x)-A(x^*)||_0/M$  and the Hamming distance between A(x)and y, defined as  $||A(x) - y||_0/M$ , is used to measure the difference between A(x) and the noisy measurements y. The results are depicted in Figure 1. The plots demonstrate that in these comparisons four AOP algorithms outperform BIHT and BIHT- $\ell_2$  for all noise levels, significantly so when more than 2% of the measurements are corrupted. Compared with BIHT, BIHT- $\ell_2$  tends to give worse results when there are only a few sign flips in y and better results if we have high noise level. This has been shown and studied in [15]. Of all the AOP series, AOP and AOP-f give better results compared with AOP- $\ell_2$  and AOP- $\ell_2$ -f. We can also see that there is a lot of overlap between the results obtained by AOP and the ones acquired by AOP with flips, especially when one-sided  $\ell_2$  objective is used, the results are almost the same. Figure 1(d) compares the average Hamming distances between A(x) and the noisy measurements y for all algorithms. If the sign flips can be found correctly, then the Hamming distance between A(x) and y should be equal to noise level. The result shows that average Hamming distances for AOP and AOP-f are slightly above the noise levels, which means that AOP with one-sided  $\ell_1$ objective performs better in consistency than other algorithms in noisy cases.

In order to show that our algorithms can find the positions of sign flips with high accuracy, we measure the probabilities of correct detections of sign flips in the noisy measurements for different noise levels from 0.5% to 10% in Figure 2 (M = N = 1000, K = 10). The exact number of sign flips is used as L in the algorithms and we compare the exact locations of sign flips in measurements y with those detected from the algorithms for all 100 trials, then the average probabilities of correct detections are shown for different algorithms at different noise levels. From this figure, we can see that all four algorithms have high accuracy in detecting the sign flips. When the noise level is low (<4%), the accuracy of AOP and AOP-f can be as high as 95%, even when the noise level is high (e.g., 10%), the accuracy of AOP and AOP-f is still above 90%. Comparing to algorithms with one-sided  $\ell_1$  objective, algorithms with one-sided  $\ell_2$  objective have lower accuracy. The accuracy for AOP- $\ell_2$  and AOP- $\ell_2$ -f is around 80%.



Fig. 2: The probabilities of correct detections of sign flips for different noise levels ranging from 0.5% to 10%. AOP and AOP-f have very high accuracy (great than 90%) in detecting the sign flips, while AOP- $\ell_2$  and AOP- $\ell_2$ -f have relatively lower accuracy (around 80%).

# B. M/N test

In the second experiment, N = 1000, K = 10 and the noise level 3% are fixed, and we change M/N within the range (0,2]. 40 different M/N are considered and we perform 300 tests for each value. The results are displayed in five different ways: the average SNR, average angular error, average Hamming error between A(x) and  $A(x^*)$ , average Hamming distance between A(x) and y and average percentage of coefficient "misses". Here "misses" stands for the coefficients where  $x_i^* \neq 0$  while  $x_i = 0$ . According to Figure 3, although all the algorithms show the same trend as M/N increases, AOP and AOP-f always obtain a much smaller angular error (higher SNR) than BIHT and BIHT- $\ell_2$ . There are also fewer coefficient misses in the results acquired by AOP series. Furthermore, we see that even when 3% of the measurements are corrupted, AOP can still recover a



Fig. 4: Hamming error vs angular error with different M. AOP gives the most consistent results for M = 0.7N and M = 1.5N. In these two cases we can see a linear relationship  $\epsilon_{sim} \approx C + \epsilon_H$  between the average angular error  $\epsilon_{sim}$  and average Hamming error  $\epsilon_H$ , where C is constant. For really small M (M = 0.1N) BIHT returns almost the same results as AOP since AOP may fail to find the exact sign flips in the noisy measurements. The dashed line  $\epsilon_{1000} + \epsilon_H$  is a upper bound for 1000 trials.



(c) Hamming distance between A(x)(d) Hamming error between A(x) and and  $A(x^*)$ y



Fig. 3: Algorithm comparison on corrupted data with different M/N. (a) average SNR vs M/N, (b) average angular error vs M/N, (c) average Hamming error between A(x) and  $A(x^*)$ vs M/N, (d) average Hamming distance between A(x) and y vs M/N, (e) average percentage of coefficient misses vs M/N. AOP yields a remarkable improvement in reducing the Hamming and angular error and achieving higher SNR.

signal with SNR greater than 20 using less than 0.5 bits per coefficient of  $x^*$ . In Hamming error comparison, AOP and AOP-f beat other algorithms significantly when M/N > 0.15. Moreover, we see that the average Hamming error of AOP and AOP-f is extremely close to zero when M/N > 0.5. When M/N < 0.15, the seemingly failure of AOP and AOPf compared with BIHT is due to the fact that there are usually more than one solution to (6) for really small M, and with high probability our method will return one solution with L sign flips, which may not be the actual ones. Hence we may not be able to detect the actual errors in the measurements.

We also try to explore the relationship between the Hamming error between A(x) and  $A(x^*)$  and the reconstruction angular error. With N = 1000, K = 10 and the noise level 3% fixed, we plot the Hamming error vs angular error for three different M in Figure 4. Since AOP and AOP with flips tend to return almost the same results if we use the same objective (one-sided  $\ell_1$  or one-sided  $\ell_2$ ) for x update, we only compare the results acquired by BIHT, BIHT-l2, AOP and AOP- $\ell_2$ . We can see clearly that almost all the blue (+) points stay in the lower left part of the graph for M = 0.7N and M = 1.5N, which proves that AOP gives more consistent results compared with other three algorithms. For these two M, the average angular error is close to a linear function of average Hamming error, which is predicted by  $B \in SE$  property in [15]. We also plot an empirical upper bound for AOP of  $\epsilon_{1000} + \epsilon_H$  defined in [15], where  $\epsilon_{1000}$  is the largest angular error of AOP and  $\epsilon_H$  is the Hamming distance. For especially "under-sampled" case like M = 0.1N, none of these algorithms is able to return consistent reconstructions, as we can see the points scatter almost randomly over the domain. In this case the results obtained by BIHT stay really close to those gained by AOP. As mentioned above, this is because AOP may not be able to detect the exact sign flips in the noisy measurements when M is too small.

## C. high noise levels

In this subsection, we study the performance of AOP and AOP- $\ell_2$  when a large number of measurements are corrupted.

Two settings are considered. In the first experiment, we fix N = 1000, K = 10, and change the M/N ratio between 0.05 and 2. Four different noise levels are considered from 0.1 to 0.4 and we record the average angular error and correct detection probability from 100 tests. In the second setting, we fix M = 2000, N = 1000 and change K from 1 to 30. Still, four noise levels are considered and the mean results from 100 tests are recorded. From Figure 5 (a) and (b), we can see similar trend for the behavior of angular error and correct detection probability as we have discovered in Figure 3. According to (c), (d), for all the noise levels the performance of these two algorithms tends to get worse as K increases. We also have another interesting discovery that when the noise level is greater than 0.2, AOP- $\ell_2$  turns out to be a better choice than AOP. This is because when the noise level is extremely high, even with outlier detection technique, lots of sign flips remain in the recovered measurements, and this new "noise level" is still relatively high. According to [15], BIHT- $\ell_2$ outperforms BIHT when the measurements contain lots of sign flips. Therefore, when the noise level is high enough, AOP- $\ell_2$ is considered as a better choice compared with AOP.



Fig. 5: AOP and AOP- $\ell_2$  performance under different noise levels. (a) average angular error vs M/N with different noise levels, (b) correct detection percentage vs M/N with different noise levels, (c) average angular error vs K with different noise levels, (d) correct detection percentage vs K with different noise levels. The performance gets better when we increase M/N or decrease K.

## D. L mismatch

In Figure 6, we analyze the influence of incorrect selection of L on AOP. Here we choose M = N = 1000, K = 10, noise level 5%, and change the input value from 0.5L to 1.5L. 100 tests are conducted and the mean results are recorded. It is easily seen that the error will become larger when the input L

digresses from its true value. According to this plot, we know that in order to obtain good performance for our method, we should choose a proper L as input.



Fig. 6: AOP performance with different L inputs. L has to stay close to its true value in order to get good performance.

## E. unknown L

To show that our method works even when L is not given, we use the method described in Section III to find an approximation of L, and compare the results of AOP with different L. Here M = N = 1000, K = 10 are fixed, and 10 different noise levels (from 1% to 10%) are tested. Three inputs for L: the initial  $L_0$  predicted from the result of BIHT- $\ell_2$ , L obtained from bisection method, exact L, are used in AOP to obtain the results. The following figure 7 is depicted with the average results from 100 trials. Even with the initial  $L_0$ , the results are comparable to those with exact L, and bisection method can provide a better approximation for Lwith longer time for predicting L.



Fig. 7: Comparison of results by different L at different noise levels from 1% to 10%. (a) average angular error vs noise level, (b) average Hamming distance between A(x) and noisy y vs noise level. By choosing appropriate L as the input, we can still obtain the results comparable to those with exact L.

## V. CONCLUSION

In this paper, we propose a method based on adaptive outlier pursuit for robust 1-bit compressive sensing. By iteratively detecting the sign flips in measurements and recovering the signals from "correct" measurements, this method can obtain better results in both finding the noisy measurements and recovering the signals, even when there are a lot of sign flips in the measurements. Four algorithms (AOP, AOP-f, AOP- $\ell_2$  and AOP- $\ell_2$ -f) are given based on this method, and the performances of these four algorithms are shown in the numerical experiments. The algorithms based on one-sided  $\ell_1$  objective (AOP and AOP-f) have better performance compared to the other two algorithm (AOP- $\ell_2$  and AOP- $\ell_2$ -f), which are based on one-sided  $\ell_2$  objective when the noise level is not high (less than 20%), when the noise level is extremely high, AOP- $\ell_2$  is a better choice compared with AOP. In addition, we proposed a simple method to find a candidate for the number of sign flips *L* when *L* is unknown and the numerical experiments show that the performance of AOP with this inexact input *L* is comparable with that of exact *L*.

## APPENDIX

In this appendix, we show the equivalence of problem (6) and (7). If (x, n) satisfies the constraints of problem (7), we can define

$$\Lambda_i = \begin{cases} 1, & \text{if } n_i = 0, \\ 0, & \text{otherwise.} \end{cases}$$
(11)

then we have  $\phi(y_i, (\Phi x)_i + n_i) = 0$  if  $\Lambda_i = 0$ , since we can always find  $n_i$  such that  $\phi(y_i, (\Phi x)_i + n_i) = 0$  for fixed x. If  $\Lambda_i = 1$ , we have  $n_i = 0$ , thus  $\phi(y_i, (\Phi x)_i + n_i) = \phi(y_i, (\Phi x)_i)$ . Therefore, problem (7) is equivalent to

$$\min_{\substack{x,n \\ s.t.}} \sum_{i=1}^{M} \Lambda_i \phi(y_i, (\Phi x)_i) \\
\|n\|_0 \le L, \\
\|x\|_2 = 1, \quad \|x\|_0 \le K.$$
(12)

From the relation of  $\Lambda$  and n in (11), we know the constraint  $||n||_0 \leq L$  in the above problem can be replaced with the constraints on  $\Lambda$  defined in (6). Therefore, problem (6) and (7) are equivalent.

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