SCIENCE CHINA

Information Sciences



• LETTER •

February 2023, Vol. 66 129202:1-129202:2 https://doi.org/10.1007/s11432-022-3546-5

A recursive least squares algorithm with ℓ_1 regularization for sparse representation

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Received 9 March 2022/Revised 8 May 2022/Accepted 17 June 2022/Published online 5 January 2023

Citation Liu D, Baldi S, Liu Q, et al. A recursive least squares algorithm with ℓ₁ regularization for sparse

Sparse representation aims to identify a few basic elements in a signal, so as to use a combination of such elements to reconstruct the original signal. The ℓ_1 -norm has been widely applied in sparse representation, either when processing batches of data offline, or online as in adaptive filtering [1-3]. Representative algorithms for online sparse representation include zero-attracting LMS (ZA-LMS) [4], ℓ_1 -RLS [5], ℓ_1 -RRLS [6], zero-attracting RLS (ZA-RLS) [7], among others [8, 9]. A new recursive least squares (RLS) method for sparse representation, named ℓ_1^2 -RLS, is proposed in this study. The method, relying on minimizing an appropriately designed cost with ℓ_1 regularization, is compared with state-of-the-art algorithms adopting ℓ_2 and ℓ_1

Proposed ℓ_1^2 -RLS method. Consider an input-output parameter estimation setting given by the standard relation:

$$y(k) = \sum_{i=0}^{N-1} h_i x_i(k) + n(k) = \mathbf{h}^{\mathrm{T}} \mathbf{x}(k) + n(k), \qquad (1)$$

where $\boldsymbol{h} = [h_0 \, h_1 \cdots h_{N-1}]^{\mathrm{T}}$ denotes the unknown weight vector to be estimated, $\mathbf{x}(k) = [x_0(k) x_1(k) \cdots x_{N-1}(k)]^T$ is the input signal, y(k) is the output signal, n(k) is the measurement noise, and k is the time index. The system is sparse when only a few elements of h are non-zero. The estimation goal is to provide an estimate of h at time k, called h(k), by using input and output signals collected up to time k.

To formulate the estimation problem, let us define the following least squares cost function with ℓ_1 -regularization:

$$J_N(k) = \frac{1}{2} \sum_{i=1}^k (y(i) - \widehat{\boldsymbol{h}}^{\mathrm{T}}(k) \boldsymbol{x}(i))^2 + \frac{\rho}{2} ||\widehat{\boldsymbol{h}}(k)||_1^2$$
(2)

$$+ \frac{1}{2} (\widehat{\boldsymbol{h}}(k) - \widehat{\boldsymbol{h}}^{\mathrm{T}}(0)) P^{-1}(0) (\widehat{\boldsymbol{h}}(k) - \widehat{\boldsymbol{h}}(0)),$$

where $\hat{h}(0)$ is an initial estimate of h, P(0) is an initial covariance matrix that weights the ℓ_2 -norm, and $\rho > 0$ is the regularizing parameter that weights the ℓ_1 -norm. Note the square of the ℓ_1 -norm in (2), consistent with the square of the ℓ_2 -norm in Tikhonov regularization. The resulting ℓ_1^2 -RLS method is in Algorithm 1. The details of the derivation of the algorithm are in Appendix A.

Algorithm 1 Proposed ℓ_1^2 regularized recursive least squares

Data and parameters: $\boldsymbol{x}(n), y(n), \rho, \, \hat{\boldsymbol{h}}(0), P(0).$

- 1: for time step $k=1,2,\ldots,n,$ do 2: Let $e(k)=y(k)-\widehat{\boldsymbol{h}}^{\mathrm{T}}(k-1)\boldsymbol{x}(k)$ and $\overline{\mathrm{sgn}}(k-1)=$ [sgn($\hat{h}_0(k-1)$) sgn($\hat{h}_1(k-1)$) \cdots sgn($\hat{h}_{N-1}(k-1)$)]^T; 3: Let $Q(k) = [\boldsymbol{x}(k) \sqrt{\rho \operatorname{sgn}}(k-1) - \sqrt{\rho \operatorname{sgn}}(k-2)]$; 4: Let $S(k) = [\boldsymbol{x}(k) \sqrt{\rho \operatorname{sgn}}(k-1) \sqrt{\rho \operatorname{sgn}}(k-2)]$ ^T; 5: Update $P(k) = P(k-1) - P(k-1)Q(k) \times (I + S(k)P(k-1)Q(k))^{-1}S(k)Q(k)$.

- 5. Optate $\widehat{h}(k) = \widehat{I}(k-1) = \widehat{I}(k-1) \cdot \widehat{I}(k) + \widehat{I}(k) \cdot \widehat{I}(k-1) \cdot \widehat{I}(k) = \widehat{I}(k-1) \cdot \widehat{I}(k) = \widehat{I}(k-1) + P(k)x(k)e(k) \rho P(k)[\overline{sgn}(k-1) \overline{sgn}(k-2)] \cdot \frac{\overline{sgn}^{T}(k-1)}{\overline{sgn}^{T}(k-2)} \widehat{I}(k-1);$
- 7: end for

Remark 1. By looking at steps 5 and 6 of Algorithm 1, one can notice that the proposed ℓ_1^2 -RLS algorithm adds a few extra terms to the standard RLS algorithm, which are

$$P(k) \, = \, P(k-1) - \, \frac{P(k-1) \boldsymbol{x}(k) \boldsymbol{x}^{\mathrm{T}}(k) P(k-1)}{1 + \boldsymbol{x}^{\mathrm{T}}(k) P(k-1) \boldsymbol{x}(k)},$$

$$\widehat{\boldsymbol{h}}(k) = \widehat{\boldsymbol{h}}(k-1) + P(k)\boldsymbol{x}(k)e(k).$$

In step 5 of Algorithm 1, new terms appear resulting from the matrix inversion lemma: the inverse in step 5 involves a 3×3 matrix I + S(k)P(k-1)Q(k) instead of a scalar inverse as in the standard RLS. In step 6, notice that the term

$$-\rho P(k)[\overline{\mathbf{sgn}}(k-1) \quad -\overline{\mathbf{sgn}}(k-2)] \begin{bmatrix} \overline{\mathbf{sgn}}^{\mathrm{T}}(k-1) \\ \overline{\mathbf{sgn}}^{\mathrm{T}}(k-2) \end{bmatrix} \widehat{\boldsymbol{h}}(k-1)$$

plays the role of attracting the estimate $\hat{h}(k)$ towards zero. In this sense, the philosophy to induce sparsity is analogous

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	Effect of regularization, $K = 3$							Effect of sparsity					
	$\rho = 0.1$	$\rho = 0.5$	$\rho = 1$	$\rho = 1.5$	$\rho = 2$	$\rho = 5$		K=1	K=3	K=5	K=7	K=9	
RLS	0.0400	0.0400	0.0400	0.0400	0.0400	0.0400		0.0400	0.0400	0.0400	0.0400	0.0400	
ℓ_1 -RRLS	0.0401	0.0400	0.0399	0.0398	0.0397	0.0391		0.0397	0.0399	0.0400	0.0401	0.0401	
ZA-RLS	0.0399	0.0397	0.0393	0.0390	0.0387	0.0370		0.0392	0.0393	0.0394	0.0397	0.0399	
ℓ_1^2 -RLS (proposed)	0.0398	0.0396	0.0392	0.0389	0.0386	0.0368		0.0396	0.0392	0.0391	0.0393	0.0397	
	Effect of regularization, $K = 5$							Effect of signal-to-noise ratio					
	$\rho = 0.1$	$\rho = 0.5$	$\rho = 1$	$\rho = 1.5$	$\rho = 2$	$\rho = 5$		SNR=1	SNR=3	SNR=5	SNR=7	SNR=10	
RLS	0.0400	0.0400	0.0400	0.0400	0.0400	0.0400		0.0503	0.0400	0.0317	0.0252	0.0179	
$\ell_1\text{-RRLS}$	0.0401	0.0401	0.0400	0.0399	0.0399	0.0395		0.0503	0.0399	0.0316	0.0252	0.0177	
ZA-RLS	0.0399	0.0397	0.0394	0.0392	0.0390	0.0378		0.0497	0.0393	0.0311	0.0249	0.0172	
ℓ_1^2 -RLS (proposed)	0.0398	0.0395	0.0391	0.0388	0.0384	0.0368		0.0496	0.0392	0.0310	0.0248	0.0172	
	Effect of regularization, $K = 7$												
	$\rho = 0.1$	$\rho = 0.5$	$\rho = 1$	$\rho = 1.5$	$\rho = 2$	$\rho = 5$							
RLS	0.0400	0.0400	0.0400	0.0400	0.0400	0.0400							
$\ell_1\text{-RRLS}$	0.0401	0.0401	0.0401	0.0400	0.0400	0.0398							
ZA-RLS	0.0399	0.0398	0.0397	0.0395	0.0394	0.0388							
ℓ_1^2 -RLS (proposed)	0.0398	0.0396	0.0393	0.0391	0.0389	0.0385							

Table 1 Summary of comparisons (the best performance is highlighted with bold notation)

to the zero-attracting methods [4,7]. When $\rho=0$, steps 5 and 6 in Algorithm 1 degenerate to the standard RLS.

Comparative experiments. The performance of the proposed ℓ_1^2 -RLS is compared with the standard RLS [1], ℓ_1 -RRLS [6], and ZA-RLS [7]. The input $\boldsymbol{x}(k)$ is white and n(k) is additive white Gaussian noise with a certain signal-to-noise ratio (SNR). The system is as in (1) with a total of 10 coefficients, where only K of them are nonzero. For every experiment, \boldsymbol{h} is normalized in so that $||\boldsymbol{h}||_1 = ||\boldsymbol{h}||_1^2 = 1$, which guarantees fair numerical comparisons as the penalty with ρ has similar effect for all algorithms. The performance of all methods is tested in three aspects.

- (1) The effect of regularization parameter ρ on the performance. This is done because increasing ρ increases the zero-attracting effect (driving the estimate towards zero). Therefore, we test the zero-attracting effect for different levels of sparsity. The SNR is 3 dB in all cases.
- (2) The effect of sparsity on the performance. We change the number of nonzero coefficients K to change the level of sparsity. The SNR is 3 dB in all cases.
- (3) The effect of signal-to-noise ratio on the performance. We change the SNR of the observation noise. The underlying system has 3 nonzero coefficients in all cases.

The estimation performance is evaluated based on the ℓ_2 -norm error with the true parameters, i.e., $||\hat{h}_{\rm FIN} - h||_2$, where $\hat{h}_{\rm FIN}$ is the estimated \hat{h} at the final iteration. The results are averaged over 1000 random trials, so as to obtain an average performance. The results are collected in Table 1. More details on the experiments and the algorithms used are in Appendix B.

Discussion on the results. The comparisons for different levels of regularization and sparsity show that the proposed method strikes a good trade-off between attracting the estimates towards zero (useful in sparse environments, e.g., K=3) and providing estimates close to the true \mathbf{h} . In fact, as the level of sparsity decreases (e.g., K=5 or K=7), attracting the estimate towards zero may increase the estimation error, since many elements of \mathbf{h} are different than zero. The proposed ℓ_1^2 -RLS outperforms the other methods in most scenarios. The comparisons for different levels of SNR demonstrate that the proposed ℓ_1^2 -RLS behaves better in all noisy situations. It is only when the SNR reaches 10 that ZA-RLS behaves as well as the proposed method. This good performance can be explained with the fact that

the proposed cost (2) allows a minimization approach analogous to recursive least squares with Tikhonov regularization, which behaves well in sparse and non-sparse scenarios. Thus, an interesting future study is to dynamically change the size of the vector to be estimated, i.e., to reduce or increase it online depending on an estimated degree of sparsity.

Acknowledgements This work was supported by National Natural Science Foundation of China (Grant No. 62073074), Key Intergovernmental Special Fund of National Key Research and Development Program (Grant No. 2021YFE0198700), and Research Fund for International Scientists (Grant No. 62150610499).

Supporting information Appendixes A and B. The supporting information is available online at info.scichina.com and link.springer.com. The supporting materials are published as submitted, without typesetting or editing. The responsibility for scientific accuracy and content remains entirely with the authors.

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