A Recursive Least-Squares Algorithm Based on the Nearest Kronecker Product Decomposition

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Outline

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- System Model
- RLS algorithm based on the nearest Kronecker product decomposition (RLS-NKP)
- Simulation Results
- Conclusions and Perspectives

Introduction

 Recursive least-square (RLS) algorithm → frequently used in system identification problems

 \rightarrow this algorithm is computationally very complex

In this work → new approach to improve the efficiency of the RLS → the impulse response decomposition based on the nearest Kronecker product

 \rightarrow low-rank approximation

• **Target:** a high-dimension system identification problem

→ RLS algorithm based on the nearest Kronecker product decomposition

System Model

Model

$$d(t) = (\mathbf{\hat{h}}^T \mathbf{x}(t) + w(t))$$
 where $d(t)$ - desired signal
 $w(t)$ - additive noise

→ **h** is the impulse response of the unknown system of length $L = L_1 L_2 (L_1 \ge L_2)$. → The impulse response can be decomposed as: $\mathbf{h} = \begin{bmatrix} \mathbf{s}_1^T & \mathbf{s}_2^T & \dots & \mathbf{s}_{L_2}^T \end{bmatrix}^T$ where $\mathbf{s}_l, l = 1, 2, \dots, L_2$ - short impulse responses (of length L_1) → ? **h** can be approximated as $\mathbf{h}_2 \otimes \mathbf{h}_1$ $(\mathbf{h}_2$ - length L_2, \mathbf{h}_1 - length L_1) → The normalized misalignment:

$$\mathcal{M}(\mathbf{h}_1, \mathbf{h}_2) = \frac{\|\mathbf{h} - \mathbf{h}_2 \otimes \mathbf{h}_1\|_2}{\|\mathbf{h}\|_2}$$

System Model

→ We can reorganize the components of **h** into a matrix: $\mathbf{H} = \begin{bmatrix} \mathbf{s}_1 & \mathbf{s}_2 & \dots & \mathbf{s}_{L_2} \end{bmatrix}$

$$\mathcal{M}(\mathbf{h}_1, \mathbf{h}_2) = \frac{\left\| \mathbf{H} - \mathbf{h}_1 \mathbf{h}_2^T \right\|_F}{\left\| \mathbf{H} \right\|_F}$$

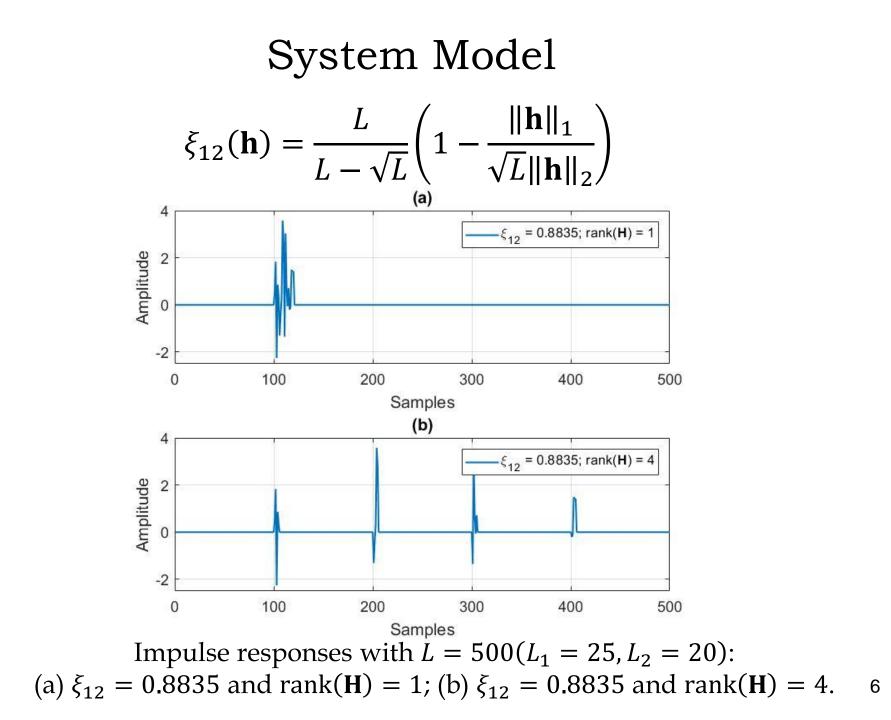
- → The optimal values of \mathbf{h}_1 and $\mathbf{h}_2 \Longrightarrow$ minimization of $\mathcal{M}(\mathbf{h}_1, \mathbf{h}_2)$
- → Minimizing $\mathcal{M}(\mathbf{h}_1, \mathbf{h}_2) \iff$ finding the nearest rank-1 matrix to **H**

$$\mathbf{H} = \mathbf{U}_{1} \Sigma \mathbf{U}_{2}^{T} = \sum_{l=1}^{L_{2}} \sigma_{l} \mathbf{u}_{1,l} \mathbf{u}_{2,l}^{T}$$

$$\mathbf{h} \approx \sum_{p=1}^{P} \mathbf{h}_{2,p} \otimes \mathbf{h}_{1,p} \quad \text{where } P \leq L_{2}$$

$$\overline{\mathbf{h}}(P) \approx \sum_{p=1}^{P} \overline{\mathbf{h}}_{2,p} \otimes \overline{\mathbf{h}}_{1,p} = \sum_{p=1}^{P} \sigma_{p} \mathbf{u}_{2,p} \otimes \mathbf{u}_{1,p}$$

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→ The goal is to estimate **h** with an adaptive filter $\hat{\mathbf{h}}(t)$ → The error signal:

$$e(t) = d(t) - \hat{y}(t) = d(t) - \hat{\mathbf{h}}^T (t - 1) \mathbf{x}(t \rightarrow We \text{ can decompose the adaptive filter:}$$

$$\hat{\mathbf{h}}(t) = \sum_{p=1}^{P} \hat{\mathbf{h}}_{2,p}(t) \otimes \hat{\mathbf{h}}_{1,p}(t)$$

$$E(t) = d(t) - \sum_{p=1}^{P} \hat{\mathbf{h}}_{1,p}^{T}(t-1) \mathbf{x}_{2,p}(t) = d(t) - \hat{\mathbf{h}}_{1}^{T}(t-1) \mathbf{x}_{2}(t)$$

$$E(t) = d(t) - \sum_{p=1}^{P} \hat{\mathbf{h}}_{2,p}^{T}(t-1) \mathbf{x}_{1,p}(t) = d(t) - \hat{\mathbf{h}}_{2}^{T}(t-1) \mathbf{x}_{1}(t)$$

$$\mathbf{x}_{2,p} = [\hat{\mathbf{h}}_{2,p}(t-1) \otimes \mathbf{I}_{L_{1}}]^{T} \mathbf{x}(t) \qquad \hat{\mathbf{h}}_{1}(t) = [\hat{\mathbf{h}}_{1,1}^{T}(t) \quad \hat{\mathbf{h}}_{1,2}^{T}(t) \qquad \dots \quad \hat{\mathbf{h}}_{1,p}^{T}(t)]^{T}$$

$$\mathbf{x}_{1,p} = [\mathbf{I}_{L_{2}} \otimes \hat{\mathbf{h}}_{1,p}(t-1)]^{T} \mathbf{x}(t) \qquad \hat{\mathbf{h}}_{2}(t) = [\hat{\mathbf{h}}_{2,1}^{T}(t) \quad \hat{\mathbf{h}}_{2,2}^{T}(t) \qquad \dots \quad \hat{\mathbf{h}}_{2,p}^{T}(t)]^{T}$$

$$\mathbf{x}_{1}(t) = [\mathbf{x}_{1,1}^{T}(t) \quad \mathbf{x}_{1,2}^{T}(t) \qquad \dots \quad \mathbf{x}_{1,p}^{T}(t)]^{T} \qquad \mathbf{x}_{2}(t) = [\mathbf{x}_{2,1}^{T}(t) \quad \mathbf{x}_{2,2}^{T}(t) \qquad \dots \quad \mathbf{x}_{2,p}^{T}(t)]^{T}$$

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→ The cost functions:

$$\begin{aligned} \mathcal{J}_{\underline{\hat{h}}_{2}}[\underline{\hat{h}}_{1}(t)] &= \sum_{i=1}^{t} \lambda_{1}^{t-i} \left[d(i) - \underline{\hat{h}}_{1}^{T}(t) \underline{\mathbf{x}}_{2}(i) \right]^{2} \\ \mathcal{J}_{\underline{\hat{h}}_{1}}[\underline{\hat{h}}_{2}(t)] &= \sum_{i=1}^{t} \lambda_{2}^{t-i} \left[d(i) - \underline{\hat{h}}_{2}^{T}(t) \underline{\mathbf{x}}_{1}(i) \right]^{2} \\ \lambda_{1}, \lambda_{2} - \text{forgetting factors} \\ \text{Normal equations} \quad \underline{\mathbf{R}}_{2}(t) \underline{\hat{h}}_{1}(t) &= \underline{\mathbf{p}}_{2}(t) \\ \underline{\mathbf{R}}_{1}(t) \underline{\hat{h}}_{2}(t) &= \underline{\mathbf{p}}_{1}(t) \end{aligned}$$
where $\underline{\mathbf{R}}_{2}(t) = \lambda_{1} \underline{\mathbf{R}}_{2}(t-1) + \underline{\mathbf{x}}_{2}(t) \underline{\mathbf{x}}_{2}^{T}(t) \quad \underline{\mathbf{p}}_{2}(t) = \lambda_{1} \underline{\mathbf{p}}_{2}(t-1) + \underline{\mathbf{x}}_{2}(t) d(t) \\ \underline{\mathbf{R}}_{1}(t) &= \lambda_{2} \underline{\mathbf{R}}_{1}(t-1) + \underline{\mathbf{x}}_{1}(t) \underline{\mathbf{x}}_{1}^{T}(t) \quad \underline{\mathbf{p}}_{1}(t) = \lambda_{2} \underline{\mathbf{p}}_{1}(t-1) + \underline{\mathbf{x}}_{1}(t) d(t) \end{aligned}$

→ The RLS-NKP: $\underline{\hat{\mathbf{h}}}_1(t) = \underline{\hat{\mathbf{h}}}_1(t-1) + \mathbf{k}_2(t)e(t)$ $\underline{\hat{\mathbf{h}}}_2(t) = \underline{\hat{\mathbf{h}}}_2(t-1) + \mathbf{k}_1(t)e(t)$

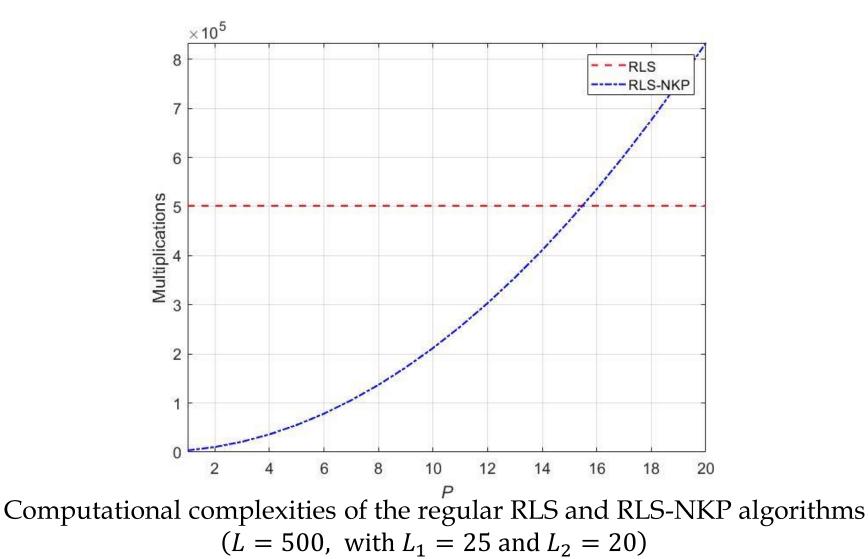
→ The Kalman gain vectors:

$$\mathbf{k}_{2}(t) = \frac{\underline{\mathbf{R}}_{2}^{-1}(t-1)\underline{\mathbf{x}}_{2}(t)}{\lambda_{1} + \underline{\mathbf{x}}_{2}^{T}(t)\underline{\mathbf{R}}_{2}^{-1}(t-1)\underline{\mathbf{x}}_{2}(t)}$$
$$\mathbf{k}_{1}(t) = \frac{\underline{\mathbf{R}}_{1}^{-1}(t-1)\underline{\mathbf{x}}_{1}(t)}{\lambda_{2} + \underline{\mathbf{x}}_{1}^{T}(t)\underline{\mathbf{R}}_{1}^{-1}(t-1)\underline{\mathbf{x}}_{1}(t)}$$

→ The updates of $\underline{\mathbf{R}}_1^{-1}(t)$ and $\underline{\mathbf{R}}_2^{-1}(t)$ (based on the inversion lemma):

$$\underline{\mathbf{R}}_{2}^{-1}(t) = \lambda_{1}^{-1} \left[\underline{\mathbf{R}}_{2}^{-1}(t-1) - \mathbf{k}_{2}(t) \underline{\mathbf{x}}_{2}^{T}(t) \underline{\mathbf{R}}_{2}^{-1}(t-1) \right]$$

$$\underline{\mathbf{R}}_{1}^{-1}(t) = \lambda_{2}^{-1} \left[\underline{\mathbf{R}}_{1}^{-1}(t-1) - \mathbf{k}_{1}(t) \underline{\mathbf{x}}_{1}^{T}(t) \underline{\mathbf{R}}_{1}^{-1}(t-1) \right]$$



Simulation Results

• conditions

- → **h** echo paths from G168 Recommendation, random impulse responses (L = 500), and an acoustic echo path (L = 1024)
- → input signals AR1(0.9) process/ speech sequence
- → additive noise w(t) WGN (SNR=20 dB)
- → measures of performance: normalized misalignment (NM)

$$\mathrm{NM}[\mathrm{dB}] = 20 \log_{10} \frac{\left\|\mathbf{h} - \hat{\mathbf{h}}(t)\right\|_{2}}{\left\|\mathbf{h}\right\|_{2}}$$

• algorithms

→ proposed RLS algorithm based on the nearest Kronecker product decomposition – <u>RLS-NKP</u>($\lambda_1 = 1 - 1/[K(PL_1)], \lambda_2 = 1 - 1/[K(PL_2)]$)

→ regular RLS [$\lambda = 1 - 1/(KL)$, with K > 1]

→ RLS-DCD [Y. V. Zakharov, *Low-Complexity RLS using dichotomous coordinate descent iterations*, IEEE Trans. Signal Process., 2008] ¹¹

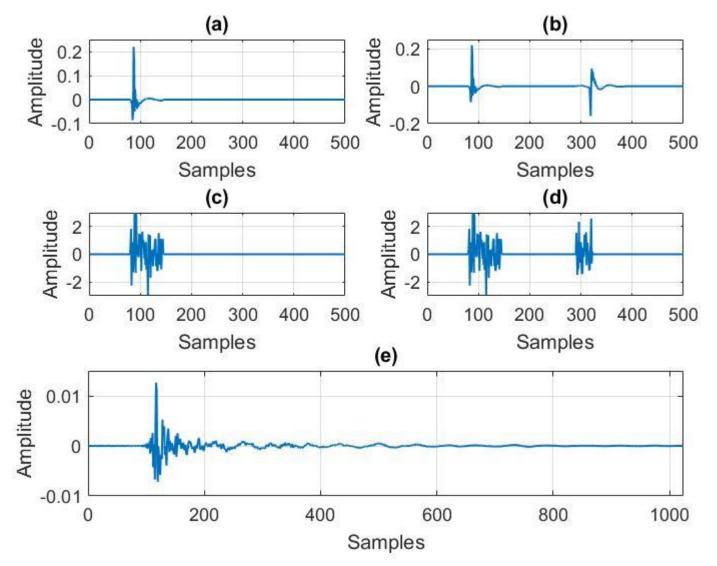


Fig. 1. Impulse responses used in the experiments: (a) $L = 500, \xi_{12} = 0.8957$, (b) $L = 500, \xi_{12} = 0.8080$, (c) $L = 500, \xi_{12} = 0.7549$, (d) $L = 500, \xi_{12} = 0.6867$, and (e) $L = 1024, \xi_{12} = 0.6880$.

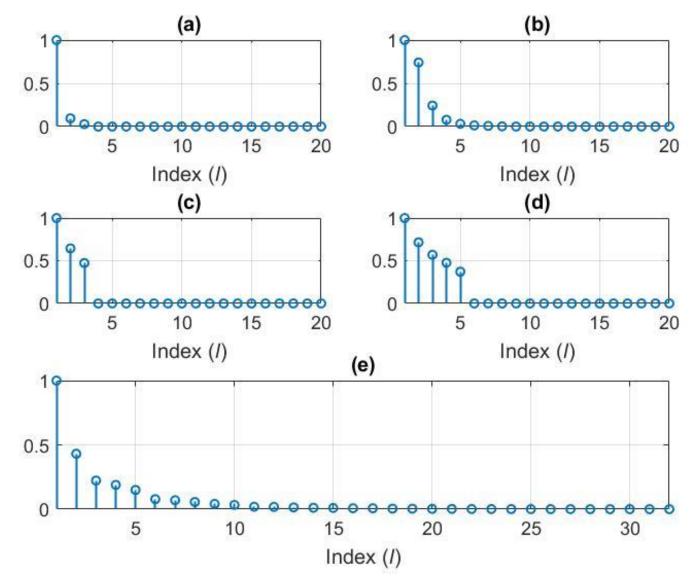


Fig. 2. Singular values (normalized with respect to the maximum one) of the matrix **H** for the corresponding impulse responses from Fig.1. The size of matrix **H** is $L_1 \times L_2$. (a)-(d) $L_1 = 25$ and $L_2 = 20$; (e) $L_1 = L_2 = 32$.

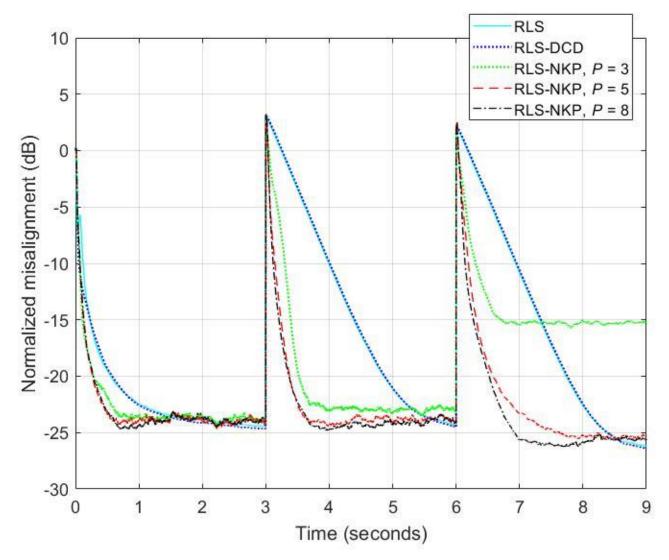


Fig. 3. Normalized misalignment of the regular RLS and RLS-DCD algorithms (L = 500), and RLS-NKP algorithm (using $L_1 = 25, L_2 = 20$, and $P < L_2$), for the identification of the impulse responses from Figs. 1(a) and (b). The input signal is an AR(1) process and the impulse response changes at times 3 and 6 seconds.

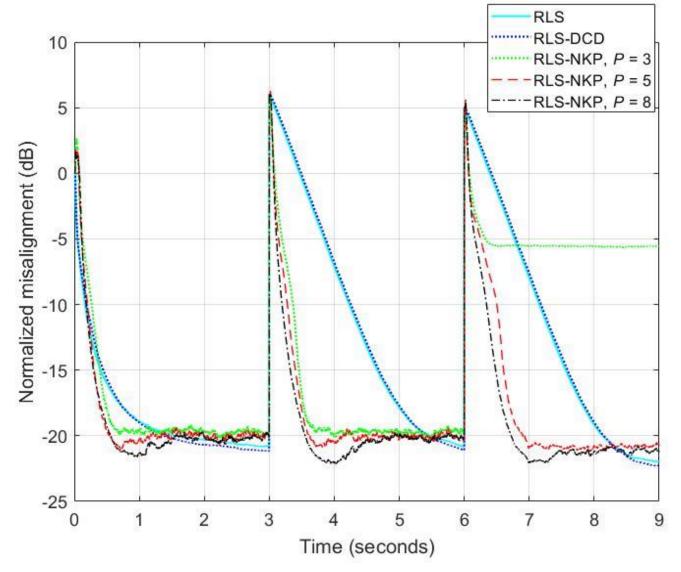


Fig. 4. Normalized misalignment of the regular RLS and RLS-DCD algorithms (L = 500), and RLS-NKP algorithm (using $L_1 = 25$, $L_2 = 20$, and P < L_2), for the identification of the impulse responses from Figs. 1(c) and (d). The input signal is an AR(1) process and the impulse response changes at times 3 and 6 seconds.

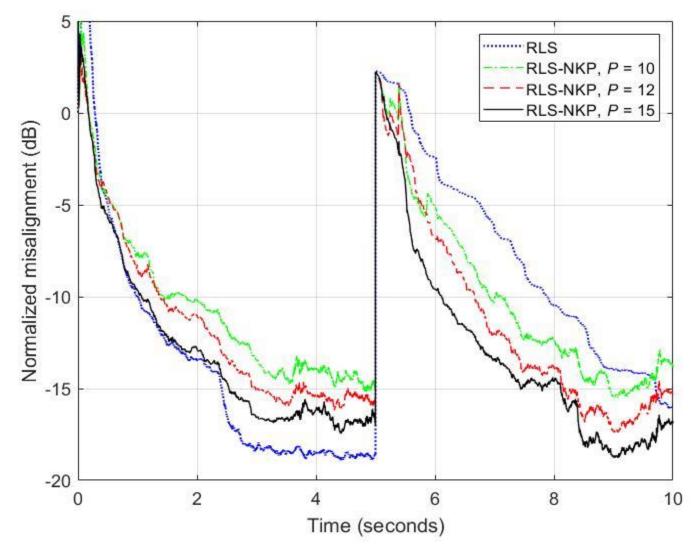


Fig. 5. Normalized misalignment of the RLS algorithm (L = 1024) and RLS-NKP algorithm (using $L_1 = L_2 = 32$, and P < L_2), for the identification of the impulse responses from Figs. 1(e). The input signal is a speech sequence and the impulse response changes at time 5 seconds.

Conclusions and Perspectives

- We have proposed the RLS-NKP algorithm.
- Suitable for the identification of low-rank models, like the echo paths.
- The tracking capabilities of the of the RLS-NKP algorithm are better as compared to the conventional RLS algorithm.
- The computational complexity of the proposed algorithm could be much lower as compared to the RLS.

C. Elisei-Iliescu, C. Paleologu, J. Benesty, C. Stanciu, C. Anghel, and S. Ciochină, "Recursive Least-Squares Algorithms for the Identification of Low-Rank Systems," *IEEE/ACM Trans. Audio, Speech, Language Process.*, May 2019.

Thank you for your attention!