

Gaussian Maximum-Likelihood Channel Estimation With Short Training Sequences

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Abstract—In this paper, we address the problem of identifying convolutive channels using a Gaussian maximum-likelihood (ML) approach when short training sequences (possibly shorter than the channel impulse-response length) are periodically inserted in the transmitted signal. We consider the case where the channel is quasi-static (i.e., the sampling period is several orders of magnitude smaller than the coherence time of the channel). Several training sequences can thus be used in order to produce the channel estimate. The proposed method can be classified as semiblind and exploits all channel-output samples containing contributions from the training sequences (including those containing contributions from the unknown surrounding data symbols). Experimental results show that the proposed method closely approaches the Cramer–Rao bound and outperforms existing training-based methods (which solely exploit the channel-output samples containing contributions from the training sequences only). Existing semiblind ML methods are tested as well and appear to be outperformed by the proposed method in the considered context. A major advantage of the proposed approach is its computational complexity, which is significantly lower than that of existing semiblind methods.

Index Terms—Block transmission, maximum-likelihood (ML) estimation, stationary multipath channel, training sequence.

I. INTRODUCTION

A MAJOR impediment of broadband communication systems is that the sampling period can become smaller than the delay spread of the channel, especially in multipath scenarios. This results in intersymbol interference (ISI), a phe-

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nomenon that needs to be combated at the receiver in order to restore the transmitted information. This is usually done using serial or block equalization techniques. Channel-state information (CSI) is needed at the receiver in order to design the equalizer and combat the ISI in an efficient way. The CSI is obtained through the use of channel-identification algorithms. These can be divided in three families that are termed “blind,” “semiblind,” and “training-based.”

Training-based techniques assume that known symbols (training or pilot symbols) are inserted in the transmitted signals. It is then possible to identify the channel at the receiver exploiting the knowledge of these known symbols (see, e.g., [1], [2], or [3]). Most existing methods require the training sequences to be significantly longer than the channel impulse response and only exploit the channel-output samples that are not corrupted by unknown data symbols. In additive-white-Gaussian-noise (AWGN) conditions, the problem of performing maximum-likelihood (ML) channel identification when only these data-free received symbols are used is then a simple least-squares problem.

Blind algorithms, on the other hand, estimate the channel based on the properties of the transmitted signals (finite alphabet properties, higher order statistics, cyclostationarity, see, e.g., [4] or [5], and references therein) and do not exploit the knowledge of possibly inserted pilots or training sequences in the stream of transmitted symbols.

Semiblind techniques have been proposed recently, which show improved performance as they simultaneously exploit the knowledge of the pilots or training symbols and blind properties of the transmitted signals towards the determination of the channel model. These techniques are most useful when the transmitted signals contain a mix of training symbols and unknown data symbols, in which case neither blind nor training-based techniques are optimal. A method that explicitly combines a blind and a training-based cost function was proposed in [6] and seems to be the first suboptimal semiblind method proposed in the literature. Later, this idea has also been adopted in [7] and [8]. A deterministic method exploiting all the energy that is received from the training sequences was presented in [9] and is suited to the case where a constant training sequence, which must be at least as long as the channel order, is periodically inserted in the transmitted sequence. Several semiblind ML channel-estimation techniques have been proposed as well. Depending on the hypothesis upon which the expression of the likelihood function is built, one can distinguish between three families of ML methods: deterministic ML, in which the data symbols are considered as deterministic disturbances, Gaussian ML, in which the data symbols are

assumed to be Gaussian distributed, and stochastic ML, where the true distribution of the data symbols is exploited. Some deterministic ML methods are presented, for instance, in [10]. In [11], a theoretical comparison of the Cramer–Rao bounds (CRBs) indicates that Gaussian ML methods outperform deterministic ML methods. In [12]–[14], Gaussian ML methods are proposed. The method presented in [14] is based on the expectation-maximization (EM) algorithm and will be used as a benchmark in this paper. Stochastic ML methods theoretically perform better as they assume a more realistic distribution for the data symbols, but suffer from a higher complexity. A method based on hidden Markov models and the EM algorithm was presented in [15], but performs significantly worse than the achievable CRB.

When facing a given transmission scheme, the choice between these three families of channel-estimation techniques depends on the way the transmitted data are formatted. When a long preamble of training symbols is inserted at the beginning of a packet of data, the training-based methods are the most attractive, as they generally allow the accurate identification of the channel at a very low computational cost. In this situation, the use of semiblind methods only marginally improves the quality of the channel estimate with a significantly higher computational cost. When there are not much training symbols or no training symbols at all, the channel cannot be estimated accurately relying on the knowledge of these training symbols and one has to rely on blind methods in order to obtain the CSI. When the number of training symbols is significant but not large enough to enable a sufficiently accurate channel estimation, or when the training symbols are not grouped in a long preamble and the classical training-based methods cannot be used efficiently, semiblind methods outperform others as they simultaneously exploit the knowledge of the training symbols and the useful properties of the transmitted signals.

In this paper, we consider the situation where short training sequences (possibly shorter than the channel impulse response length) are repeatedly inserted between blocks of data symbols, which corresponds to known-symbol-padding (KSP) transmission [16] or pilot-symbol-assisted modulation (PSAM) [17]. We consider multipath single-input single-output (SISO) channels with a relatively large coherence time (the channel stays constant during the transmission of several blocks of data). In the presented work, we aim at finding channel-estimation strategies that are suited to this specific context. Given the placement of the training symbols in the considered framework, semiblind channel-estimation strategies seem most appropriate to obtain accurate channel models. We propose a new Gaussian ML method for semiblind channel identification that is well suited to this situation. The proposed method is able to cope with arbitrarily short training sequences (possibly shorter than the channel impulse-response length) and performs channel estimation exploiting all the received symbols that contain contributions from the training sequences. The proposed method asymptotically achieves the CRB, has a small computational complexity, and seems to outperform existing semiblind methods in a similar context. For the sake of simplicity, we consider all the training sequences to have the same length, but it is straightforward to adapt the method to

the more general case of training sequences of variable length. We consider quasi-static channels (the channel stays constant during the transmission of several blocks of data), which allows us to use several training sequences to construct the channel estimate. We investigate both the situation where the same training sequence is repeated after each block of data and the situation where the training sequence is changed after each block of data.

The structure of the paper is as follows. In Section II, we present our data model. In Section III, we derive an expression for the Gaussian likelihood function of a channel estimate and introduce some approximations which we have to rely on in order to derive low-complexity ML channel estimates. In Section IV, we show that the proposed approximations have a negligible impact on the achievable performance of ML channel estimation through a CRB analysis. We then propose an iterative algorithm that converges to the ML channel estimate (Section V). We next derive an approximate closed-form expression of the ML channel estimate, both for a constant (Section VI-A) and for a changing training sequence (Section VI-B). We then discuss the identifiability conditions of the proposed methods (Section VI-C). We experimentally test the proposed methods and compare them with existing ML methods in Section VIII, and finally draw a conclusion in Section IX.

Notation: We use upper (lower) case boldface letters to denote matrices (column vectors). \mathbf{I}_N is the identity matrix of size $N \times N$ and $\mathbf{0}_{M \times N}$ is the all-zero matrix of size $M \times N$. The operator $(\cdot)^*$ denotes the complex conjugate, and $\text{Re}(\cdot)$, the real part, and $\text{Im}(\cdot)$, the imaginary part, of a complex number. The superscript $(\cdot)^T$ denotes the transpose of a matrix and $(\cdot)^H$, the complex conjugate transpose. Finally, $\text{tr}(\cdot)$ denotes the trace of a matrix, $|\cdot|$ its determinant, and $\mathbf{A}(i, j)$ denotes the i th element of the j th column of \mathbf{A} .

II. DATA MODEL

We consider a finite impulse response (FIR) convolutive channel of order L : $\mathbf{h} = [h[0] \cdots h[L]]^T$. A burst $x[n]$, $n = 1, \dots, N$, of symbols is transmitted over the channel. Considering that the coherence time of the channel is larger than the duration of the transmitted burst, the received sequence $y[n]$ is the linear convolution of the transmitted sequence with the channel impulse response

$$y[n] = \sum_{i=0}^L h[i]x[n-i] + \eta[n] \quad (1)$$

where $\eta[n]$ is the AWGN at the receiver.

A total number of K training sequences is inserted in the burst. The k th training sequence, $\mathbf{t}_k = [t_k[1] \cdots t_k[n_t]]^T$, starts at position n_k : $[x[n_k] \cdots x[n_k + n_t - 1]]^T = \mathbf{t}_k$. Two possibilities are considered in the text: either the same training sequence is repeated after each block of data (constant-training-sequence case), or the training sequence is changed after each block (changing-training-sequence case). Define the vector \mathbf{u}_k of received symbols that contain a contribution from the k th

Since the traces in (7) and (8) are always positive, the CRB of the parametric ML problem will always be tighter than the CRB of its nonparametric counterpart. However, numerical evaluations of the CRB in realistic situations show that the impact of these trace terms is negligible: the relative difference between the two CRBs (i.e., (parametric CRB – nonparametric CRB)/parametric CRB) is less than 10^{-4} for experimental setups similar to the ones that are used in Section VIII. We can thus safely work under the proposed hypothesis and consider the solutions we will obtain in this framework as the true ML channel estimates. The nonparametric CRB can be evaluated as

$$\mathcal{J}(\mathbf{h})^{-1} = \left(\sum_{k=1}^K \mathbf{T}_k^H \mathbf{Q}^{-1} \mathbf{T}_k \right)^{-1}. \quad (11)$$

Note that this bound depends both on the channel realization (through the covariance matrix \mathbf{Q}) and on the chosen training sequences (through the training sequence matrices \mathbf{T}_k).

V. ITERATIVE PROCEDURE

When a minimization problem is separable in its variables, a common approach to find the solution is an iterative one. One iteration consists of analytically minimizing the cost function with respect to one variable while keeping the other(s) fixed. The variable with respect to which the cost function is minimized is changed in each iteration (see, e.g., [19], where this approach is used to jointly estimate the transmitted data symbols and the channel). This procedure converges to a minimum of the cost function. If the cost function is convex, this convergence point is the global minimum. If the cost function is not convex, the convergence point is the global minimum only if the starting point is sufficiently close to the global minimum.¹ In the sequel, we apply this approach to the likelihood function of the system, which leads to the ML estimate of \mathbf{Q} and \mathbf{h} . The convergence properties in this context are discussed at the end of this section.

Assume that at the i th iteration, an estimate $\hat{\mathbf{Q}}^i$ of the covariance matrix \mathbf{Q} is available. We first seek the channel estimate $\hat{\mathbf{h}}^i$ that minimizes the cost function (5) with respect to \mathbf{h} for a fixed $\mathbf{Q} = \hat{\mathbf{Q}}^i$, i.e., we compute $\hat{\mathbf{h}}^i = \mathbf{h}_{\text{ML}}(\hat{\mathbf{Q}}^i)$, where $\mathbf{h}_{\text{ML}}(\mathbf{Q}) = \arg \min_{\mathbf{h}} -\mathcal{L}$. The solution to this optimization problem can be computed as

$$\mathbf{h}_{\text{ML}}(\mathbf{Q}) = \left(\sum_{k=1}^K \mathbf{T}_k^H \mathbf{Q}^{-1} \mathbf{T}_k \right)^{-1} \sum_{k=1}^K \mathbf{T}_k^H \mathbf{Q}^{-1} \mathbf{u}_k. \quad (12)$$

We then seek the covariance matrix $\hat{\mathbf{Q}}^{i+1}$ that minimizes (5) with respect to \mathbf{Q} for a fixed $\mathbf{h} = \hat{\mathbf{h}}^i$: $\hat{\mathbf{Q}}^{i+1} = \mathbf{Q}_{\text{ML}}(\hat{\mathbf{h}}^i)$,

where $\mathbf{Q}_{\text{ML}}(\mathbf{h}) = \arg \min_{\mathbf{Q}} -\mathcal{L}$, the solution to this optimization problem can be computed as (see, e.g., [20])

$$\mathbf{Q}_{\text{ML}}(\mathbf{h}) = K^{-1} \sum_{k=1}^K (\mathbf{u}_k - \mathbf{T}_k \mathbf{h})(\mathbf{u}_k - \mathbf{T}_k \mathbf{h})^H. \quad (13)$$

$\hat{\mathbf{Q}}^{i+1}$ is then used as a starting point for the next iteration. The procedure is stopped when there is no significant difference between the estimates produced by two consecutive iterations.

Assessing the convergence properties of this iterative procedure is not straightforward in this context as the likelihood function has not been proven as convex. Moreover, it is not straightforward either to assess the possible local minima of the likelihood function. Hence, it is hardly possible to derive theoretical conditions under which a given starting point would yield a guaranteed convergence of the iterative procedure to the global minimum.

Experimental results presented in [21] show however that when the iterative procedure is initialized with a simple least-squares (LS) channel estimate, the convergence point is systematically equal to the global minimum. Initializing the iterative procedure with the LS channel estimate is equivalent to choosing $\hat{\mathbf{Q}}^0 = \mathbf{I}$. In that case, it is possible to show analytically that the channel estimate $\hat{\mathbf{h}}^1$ obtained after two iterations is a good approximation of the closed-form ML channel estimates that are derived in the next section. Hence, after two iterations, the iterative scheme proceeds from a channel estimate that lies in the immediate vicinity of the global minimum, which ensures the convergence of the iterative procedure to the true ML channel estimate when $\hat{\mathbf{Q}}^0 = \mathbf{I}$ is chosen as starting point.

VI. CLOSED-FORM SOLUTION

An alternative strategy to the iterative procedure described above consists of directly finding an analytical expression for the global minimum of the likelihood function (5). The separability property of the cost function can be exploited again in order to find this global minimum. The idea is to analytically minimize the cost function with respect to one variable. This minimum is a function of the other variable. The first variable can then be eliminated in the original cost function, which then becomes a single variable expression. When the problem is separable in its two variables, minimizing this new expression of the cost function with respect to the only variable left yields the global minimum (see, e.g., [22]). This approach is often used in ML problems and is known as the process of compressing or concentrating the likelihood function onto one variable (see, e.g., [20] or [23]). In this section, we concentrate the likelihood function on the variable \mathbf{h} in order to derive the closed-form channel estimate.

First observe that the likelihood function (5) can be expressed as

$$-\mathcal{L} = K \ln |\mathbf{Q}| + \text{tr} \left(\mathbf{Q}^{-1} \sum_{k=1}^K (\mathbf{u}_k - \mathbf{T}_k \mathbf{h})(\mathbf{u}_k - \mathbf{T}_k \mathbf{h})^H \right). \quad (14)$$

¹The convergence to the global minimum is guaranteed iff the cost function evaluated at the starting point of the iterations is lower than any local minimum, in which case the trajectory of the iterative procedure is limited to the convex region around the global minimum.

We first minimize this cost function with respect to \mathbf{Q} leading to $\mathbf{Q}_{\text{ML}}(\mathbf{h})$ as given by (13). Replacing \mathbf{Q} by $\mathbf{Q}_{\text{ML}}(\mathbf{h})$ in (14) leaves us with the following expression of the concentrated likelihood function

$$-\mathcal{L} = K \ln \left| K^{-1} \sum_{k=1}^K (\mathbf{u}_k - \mathbf{T}_k \mathbf{h})(\mathbf{u}_k - \mathbf{T}_k \mathbf{h})^H \right| + K \text{tr}(\mathbf{I}).$$

The ML channel estimate is thus computed as

$$\mathbf{h}_{\text{ML}} = \arg \min_{\mathbf{h}} \left| \sum_{k=1}^K (\mathbf{u}_k - \mathbf{T}_k \mathbf{h})(\mathbf{u}_k - \mathbf{T}_k \mathbf{h})^H \right|. \quad (15)$$

At this point, we need to distinguish between the constant-training-sequence case and the changing-training-sequence case.

A. Constant Training Sequence

In order to indicate that the training sequence after each block is the same, we simply omit the block index (subscript k) for the vector \mathbf{t} and the matrix \mathbf{T} .

Define

$$\hat{\mathbf{R}} \triangleq K^{-1} \sum_{k=1}^K \mathbf{u}_k \mathbf{u}_k^H \quad \bar{\mathbf{u}} \triangleq K^{-1} \sum_{k=1}^K \mathbf{u}_k \quad \hat{\mathbf{Q}} \triangleq \hat{\mathbf{R}} - \bar{\mathbf{u}} \bar{\mathbf{u}}^H \quad (16)$$

where $\hat{\mathbf{Q}}$ is assumed to be positive definite (a necessary condition for this to hold is $K \geq n_t + L$). Using these definitions, the matrix in the minimization problem (15) can be reexpressed as

$$\begin{aligned} K^{-1} \sum_{k=1}^K (\mathbf{u}_k - \mathbf{T} \mathbf{h})(\mathbf{u}_k - \mathbf{T} \mathbf{h})^H \\ = \hat{\mathbf{Q}} \left(\mathbf{I} + \hat{\mathbf{Q}}^{-1} (\mathbf{T} \mathbf{h} - \bar{\mathbf{u}})(\mathbf{T} \mathbf{h} - \bar{\mathbf{u}})^H \right). \end{aligned}$$

Keeping in mind that $\hat{\mathbf{Q}}$ is positive definite, our minimization problem (15) is thus equivalent to

$$\mathbf{h}_{\text{ML}} = \arg \min_{\mathbf{h}} \left| \mathbf{I} + \hat{\mathbf{Q}}^{-1} (\mathbf{T} \mathbf{h} - \bar{\mathbf{u}})(\mathbf{T} \mathbf{h} - \bar{\mathbf{u}})^H \right|. \quad (17)$$

It can be shown that

$$\begin{aligned} \left| \mathbf{I} + \hat{\mathbf{Q}}^{-1} (\mathbf{T} \mathbf{h} - \bar{\mathbf{u}})(\mathbf{T} \mathbf{h} - \bar{\mathbf{u}})^H \right| \\ = 1 + (\mathbf{T} \mathbf{h} - \bar{\mathbf{u}})^H \hat{\mathbf{Q}}^{-1} (\mathbf{T} \mathbf{h} - \bar{\mathbf{u}}). \end{aligned}$$

Hence, the minimization problem (17) is equivalent to

$$\mathbf{h}_{\text{ML}} = \arg \min_{\mathbf{h}} (\mathbf{T} \mathbf{h} - \bar{\mathbf{u}})^H \hat{\mathbf{Q}}^{-1} (\mathbf{T} \mathbf{h} - \bar{\mathbf{u}}).$$

The solution is obtained by nulling the partial derivative of this expression with respect to \mathbf{h}^H , which yields

$$\mathbf{h}_{\text{ML}} = (\mathbf{T}^H \hat{\mathbf{Q}}^{-1} \mathbf{T})^{-1} (\mathbf{T}^H \hat{\mathbf{Q}}^{-1} \bar{\mathbf{u}}). \quad (18)$$

This ML channel estimate is easy to compute and also intuitively quite appealing for it shows that the ML channel estimate is simply a fit of $\mathbf{T} \mathbf{h}$ to $\bar{\mathbf{u}}$ in a weighted-least-squares sense.

B. Changing Training Sequence

Let us first introduce the following notations

$$\begin{aligned} \mathbf{h}_{\text{LS}} &\triangleq \left(\sum_{k=1}^K \mathbf{T}_k^H \mathbf{T}_k \right)^{-1} \sum_{k=1}^K \mathbf{T}_k^H \mathbf{u}_k \\ \mathbf{g} &\triangleq \mathbf{h} - \mathbf{h}_{\text{LS}}, \\ \mathbf{g}_{\text{ML}} &\triangleq \mathbf{h}_{\text{ML}} - \mathbf{h}_{\text{LS}} \\ \mathbf{e}_k &\triangleq \mathbf{u}_k - \mathbf{T}_k \mathbf{h}_{\text{LS}} \\ \hat{\mathbf{Q}}' &\triangleq K^{-1} \sum_{k=1}^K \mathbf{e}_k \mathbf{e}_k^H \end{aligned} \quad (19)$$

where $\hat{\mathbf{Q}}'$ is assumed to be positive definite (a necessary condition therefore is $K \geq n_t + L$). Using these notations, the minimization problem (15) can be rephrased as

$$\mathbf{g}_{\text{ML}} = \arg \min_{\mathbf{g}} \left| \mathbf{I} + \hat{\mathbf{Q}}'^{-1} K^{-1} \sum_{k=1}^K [\mathbf{T}_k \mathbf{g} \mathbf{g}^H \mathbf{T}_k^H - \mathbf{T}_k \mathbf{g} \mathbf{e}_k^H - \mathbf{e}_k \mathbf{g}^H \mathbf{T}_k^H] \right|. \quad (20)$$

When K is large, both \mathbf{h}_{LS} and \mathbf{h}_{ML} are close to the true \mathbf{h} (this hypothesis is confirmed by the experimental results presented in Section VIII). We can thus assume that \mathbf{g} , and consequently, the second term in (20), is small in the vicinity of the solution. It is well known that, for $\|\Delta\| \ll 1$, $|\mathbf{I} + \Delta| \approx 1 + \text{tr}(\Delta)$. Exploiting this result together with the permutation property of the trace of a product, the ML problem can be approximated by

$$\begin{aligned} \hat{\mathbf{g}}_{\text{ML}} = \arg \min_{\mathbf{g}} \sum_{k=1}^K \left[\mathbf{g}^H \mathbf{T}_k^H \hat{\mathbf{Q}}'^{-1} \mathbf{T}_k \mathbf{g} \right. \\ \left. - \mathbf{e}_k^H \hat{\mathbf{Q}}'^{-1} \mathbf{T}_k \mathbf{g} - \mathbf{g}^H \mathbf{T}_k^H \hat{\mathbf{Q}}'^{-1} \mathbf{e}_k \right]. \end{aligned} \quad (21)$$

The solution to this minimization problem is given as

$$\hat{\mathbf{g}}_{\text{ML}} = \left(\sum_{k=1}^K \mathbf{T}_k^H \hat{\mathbf{Q}}'^{-1} \mathbf{T}_k \right)^{-1} \sum_{k=1}^K \mathbf{T}_k^H \hat{\mathbf{Q}}'^{-1} \mathbf{e}_k.$$

Replacing \mathbf{h} by $\mathbf{g} + \mathbf{h}_{\text{LS}}$ and \mathbf{e}_k by $\mathbf{u}_k - \mathbf{T}_k \mathbf{h}_{\text{LS}}$, and rearranging the resulting expression, we obtain the following approximation $\hat{\mathbf{h}}_{\text{ML}}$ of the true ML channel estimate \mathbf{h}_{ML} :

$$\hat{\mathbf{h}}_{\text{ML}} = \left(\sum_{k=1}^K \mathbf{T}_k^H \hat{\mathbf{Q}}'^{-1} \mathbf{T}_k \right)^{-1} \sum_{k=1}^K \mathbf{T}_k^H \hat{\mathbf{Q}}'^{-1} \mathbf{u}_k. \quad (22)$$

Note that this solution is obtained after two steps of the previously proposed iterative procedure.

C. Identifiability Conditions

In theory (see, e.g., [24] for a broader discussion on channel identifiability), the channel is identifiable in the considered Gaussian problem formulation when two conditions are fulfilled: the number of channel-output samples considered for channel identification is larger than the channel order L and there is at least one nonzero training symbol not located at the edges of the considered burst.

The first condition is always fulfilled in the considered data model and the second is fulfilled as soon as there is a nonzero element in each training sequence.

However, as we need to invert the estimate of the matrix \mathbf{Q} in the proposed methods (both the iterative method and the closed-form solutions), this estimate has to have a full rank. Relying on the randomness of the noise and the unknown data symbols, this happens with probability 1 as soon as $K \geq n_t + L$. Hence, the identifiability conditions of the proposed method are summarized as follows.

- 1) $n_t \geq 1$.
- 2) $K \geq n_t + L$.

VII. ASYMPTOTIC PROPERTIES OF THE CLOSED-FORM CHANNEL ESTIMATES

In this section, we study the asymptotic properties of the proposed (approximate) closed-form ML channel estimates, that is, their properties when the number of transmitted data blocks K is large.

A. Constant Training

Keeping in mind that $\mathbf{u}_k = \mathbf{T}\mathbf{h} + \epsilon_k$, $\hat{\mathbf{Q}}$ can be rewritten as

$$\hat{\mathbf{Q}} = K^{-1} \sum_{k=1}^K \epsilon_k \epsilon_k^H - K^{-2} \sum_{i,j=1}^K \epsilon_i \epsilon_j^H. \quad (23)$$

This expression shows that $\hat{\mathbf{Q}}$ is clearly a consistent estimate of \mathbf{Q}

$$\lim_{K \rightarrow \infty} \hat{\mathbf{Q}} = \mathbf{Q}. \quad (24)$$

It follows that the ML channel estimate (18) is consistent as well:

$$\lim_{K \rightarrow \infty} \mathbf{h}_{\text{ML}} = \lim_{K \rightarrow \infty} K^{-1} \sum_{k=1}^K (\mathbf{T}^H \hat{\mathbf{Q}}^{-1} \mathbf{T})^{-1} \mathbf{T}^H \hat{\mathbf{Q}}^{-1} \mathbf{u}_k = \mathbf{h}. \quad (25)$$

Relying on these results, we can derive the asymptotic efficiency of the proposed channel estimate

$$\begin{aligned} & \lim_{K \rightarrow \infty} \mathbb{E} \{ K (\mathbf{h}_{\text{ML}} - \mathbf{h}) (\mathbf{h}_{\text{ML}} - \mathbf{h})^H \} \\ &= \mathbb{E} \left\{ \lim_{K \rightarrow \infty} (\mathbf{T}^H \hat{\mathbf{Q}}^{-1} \mathbf{T})^{-1} \mathbf{T}^H \hat{\mathbf{Q}}^{-1} K^{-1} \right. \\ & \quad \left. \times \sum_{i=1}^K \epsilon_i \sum_{j=1}^K \epsilon_j^H \hat{\mathbf{Q}}^{-1} \mathbf{T} (\mathbf{T}^H \hat{\mathbf{Q}}^{-1} \mathbf{T})^{-1} \right\} \\ &= (\mathbf{T}^H \mathbf{Q}^{-1} \mathbf{T})^{-1}. \end{aligned} \quad (26)$$

This last expression is equal to the normalized CRB [see (11)].

B. Changing Training

First note that it is possible to show (the derivation is detailed in the Appendix)

$$\lim_{K \rightarrow \infty} \hat{\mathbf{Q}}' = \mathbf{Q}. \quad (27)$$

Replacing \mathbf{u}_k by its equivalent $\mathbf{T}_k \mathbf{h} + \epsilon_k$ in (22) yields

$$\hat{\mathbf{h}}_{\text{ML}} = \mathbf{h} + \left(\sum_{k=1}^K \mathbf{T}_k^H \hat{\mathbf{Q}}'^{-1} \mathbf{T}_k \right)^{-1} \sum_{k=1}^K \mathbf{T}_k^H \hat{\mathbf{Q}}'^{-1} \epsilon_k. \quad (28)$$

Using this expression of $\hat{\mathbf{h}}_{\text{ML}}$ highlights the fact that the proposed approximate ML channel estimate is consistent: $\lim_{K \rightarrow \infty} \hat{\mathbf{h}}_{\text{ML}} = \mathbf{h}$.

Here also, it is possible to derive the asymptotic efficiency of the proposed channel estimate

$$\begin{aligned} & \lim_{K \rightarrow \infty} \mathbb{E} \left\{ K (\hat{\mathbf{h}}_{\text{ML}} - \mathbf{h}) (\hat{\mathbf{h}}_{\text{ML}} - \mathbf{h})^H \right\} \\ &= \mathbb{E} \left\{ \lim_{K \rightarrow \infty} \left(K^{-1} \sum_{k=1}^K \mathbf{T}_k^H \hat{\mathbf{Q}}'^{-1} \mathbf{T}_k \right)^{-1} \right. \\ & \quad \left. \times K^{-1} \sum_{k=1}^K \mathbf{T}_k^H \hat{\mathbf{Q}}'^{-1} \epsilon_k \sum_{k=1}^K \epsilon_k^H \hat{\mathbf{Q}}'^{-1} \mathbf{T}_k \right. \\ & \quad \left. \times \left(K^{-1} \sum_{k=1}^K \mathbf{T}_k^H \hat{\mathbf{Q}}'^{-1} \mathbf{T}_k \right)^{-1} \right\} \\ &= \lim_{K \rightarrow \infty} \left(K^{-1} \sum_{k=1}^K \mathbf{T}_k^H \mathbf{Q}^{-1} \mathbf{T}_k \right)^{-1} \mathbb{E} \{ \mathbf{T}_k^H \mathbf{Q}^{-1} \mathbf{T}_k \} \\ & \quad \times \left(K^{-1} \sum_{k=1}^K \mathbf{T}_k^H \mathbf{Q}^{-1} \mathbf{T}_k \right)^{-1} \\ &= \lim_{K \rightarrow \infty} \left(K^{-1} \sum_{k=1}^K \mathbf{T}_k^H \mathbf{Q}^{-1} \mathbf{T}_k \right)^{-1}. \end{aligned}$$

The above expression is equal to the normalized CRB [see (11)].

VIII. EXPERIMENTAL RESULTS

The performance metric that is used throughout this section is the normalized mean square error (NMSE) of the proposed channel estimate

$$\text{NMSE} = \mathbb{E} \left\{ \frac{\|\hat{\mathbf{h}} - \mathbf{h}\|^2}{\|\mathbf{h}\|^2} \right\}.$$

The results that are presented are obtained with the closed-form channel estimates (18) and (22). When the iterative method results are investigated, we explicitly state it in the text. We use the CRB as a benchmark in the experiments. The CRB curves displayed on the graphs represent the NMSE of an estimator

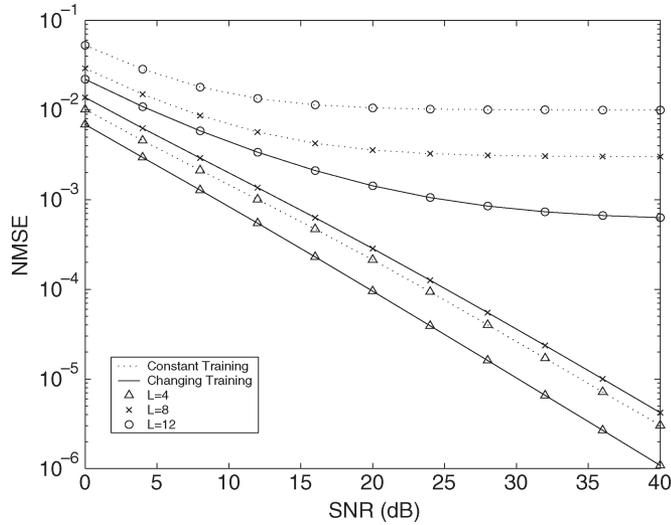


Fig. 1. Comparison of the CRB of the constant-and-changing-training-sequence cases versus the SNR for different channel orders when $n_t = 10$ and $K = 100$.

that achieves the CRB, which is $E\{\text{tr}(\mathcal{J}^{-1})/\|\mathbf{h}\|^2\}$. The experiments are performed on convolutive Rayleigh fading channels of varying order L . The different channel taps are independent identically distributed Gaussian random variables. The training and data sequences are randomly picked white QPSK sequences. The energy of the transmitted symbols (both data and training) is set to $\lambda^2 = 1$. The presented results are obtained after averaging over a set of 100 channel realizations. For each of these channel realizations, the results are averaged over 100 different sets of training sequences in the changing-training-sequence case and over 100 different training sequences in the constant-training-sequence case. Note that this averaging is also done for the CRB results, since the CRB depends both on the channel realization and the training sequences. The signal-to-noise ratio (SNR) is defined as $\text{SNR} = E\{\|\mathbf{h}\|^2\}(\lambda^2/\sigma^2)$.

A. Performance of the Proposed Method

In this section, we analyze and compare the algorithms proposed for the two situations that have been considered throughout this article: the constant and changing training-sequence cases.

1) *Comparison of the CRBs*: To have a first insight in how these compare, we check the CRB performance for these two configurations. We consider a transmission scheme where the length of the training sequences is set to $n_t = 10$ and the number of observed training sequences is set to $K = 100$. The CRB for different channel orders in that context is presented as a function of the SNR in Fig. 1. We see that the use of changing training sequences systematically results in a reduced CRB for all channel orders. In both situations, the CRB decreases with a constant slope as the SNR increases when there is an exact solution to the channel-identification problem in the noiseless case, i.e., when $n_t \geq 2L + 1$ when constant training sequences are used and when $n_t \geq L + 1$ when changing training sequences are used. The CRB saturates at high SNR, if there is no exact solution to the channel-

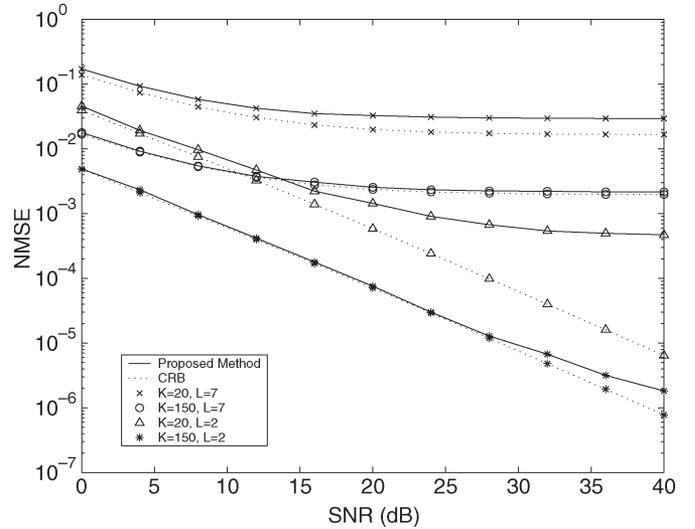


Fig. 2. Comparison of the simulated NMSE and the CRB versus SNR for different channel orders when $n_t = 5$ and changing training sequences are used. The results are plotted for two different values of K , namely 20 and 150.

identification problem in the noiseless case, i.e., when $n_t < 2L + 1$ for constant training sequences and $n_t < L + 1$ for changing training sequences. When n_t is fixed and the channel order is in the interval $((n_t - 1)/2) \leq L \leq n_t - 1$, using changing training sequences will yield a constant slope in the CRB for increasing SNR, whereas a floor will appear at high SNR if a constant training sequence is used. For channel orders outside this interval, both methods show a similar behavior (constant slope for small channel orders and saturation for large channel orders), but there is still an advantage in using changing training sequences.

2) *Changing Training Sequences*: After this discussion on the CRB, we check how the proposed closed-form channel estimates match this theoretical bound. We first check it for the closed-form ML channel estimate proposed in the context of changing training sequences. In Fig. 2, we compare the simulated performance of our method with the corresponding CRB as a function of the SNR. We perform this comparison for two different channel orders: one for which the CRB has a constant slope, the other being large enough to have the CRB saturating at high SNR. We repeat these experiments for two different values of K : a relatively small one and a larger one. When the channel order is large (saturation of the CRB at high SNRs), we observe a relatively good match between the CRB and the experimental curves. The match is tighter for a larger value of K . When the channel order is small (constant slope in the CRB curves), the theoretical and experimental curves match quite well when K is large, but we see the emergence of a floor on the experimental NMSE for higher SNR when K is small. In Fig. 3, we evaluate the impact of the number of data blocks K on the channel estimate NMSE. The simulations are done for two different channel orders and two different values of the SNR. The presented results confirm the asymptotical efficiency of the closed-form channel estimate as the experimental performance systematically achieves the CRB for large values of K . When K gets small, the match between the CRB and the experimental results remains acceptable, except

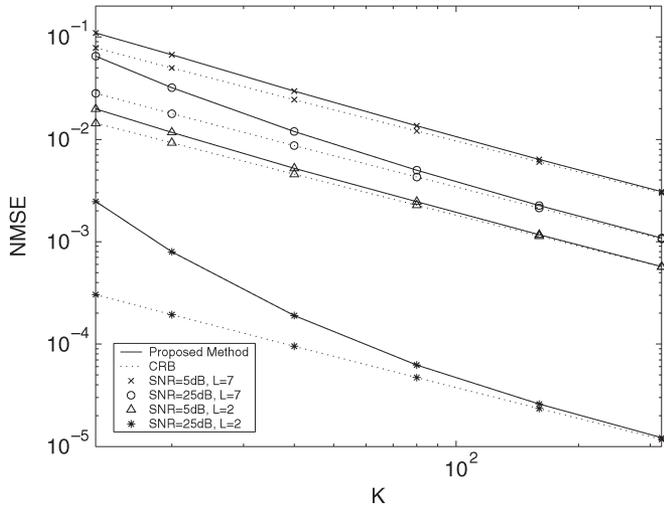


Fig. 3. Comparison of the simulated NMSE and the CRB versus K for different channel orders when $n_t = 5$ and changing training sequences are used. The results are plotted for two different values of the SNR, namely 5 and 25 dB.

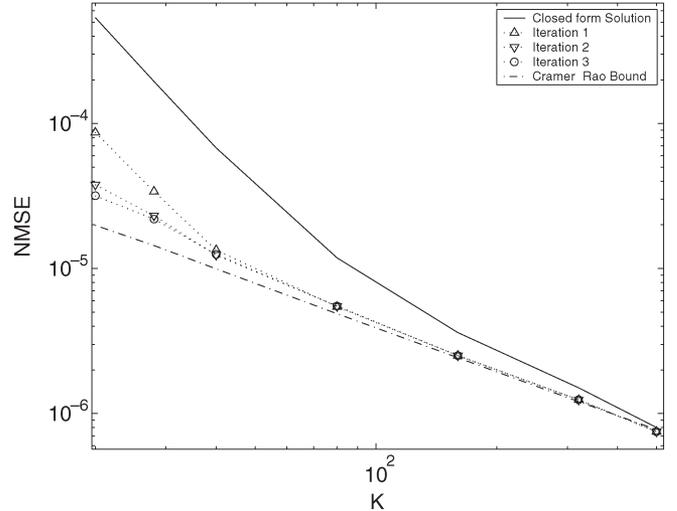


Fig. 5. Comparison of the simulated NMSE and the CRB versus K for $L = 2$, $n_t = 5$, and SNR = 35 dB using changing training sequences. The NMSE converges to the CRB after a few iterations when the iterative method is used.

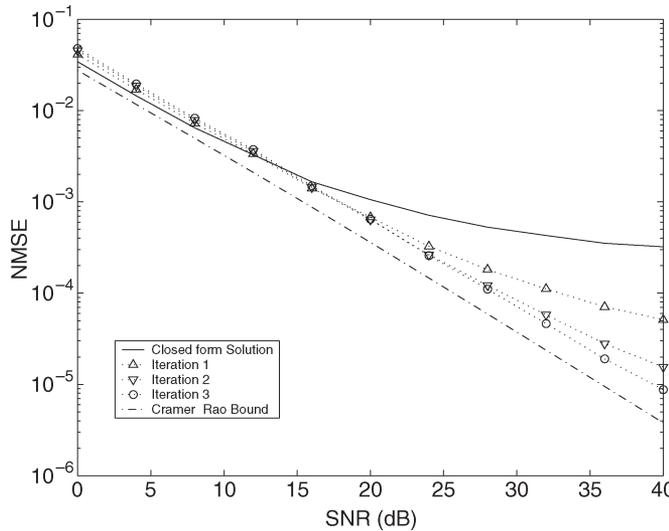


Fig. 4. Comparison of the simulated NMSE and the CRB versus SNR for $L = 2$, $n_t = 5$, and $K = 20$ using changing training sequences. The NMSE converges to the CRB after a few iterations when the iterative method is used.

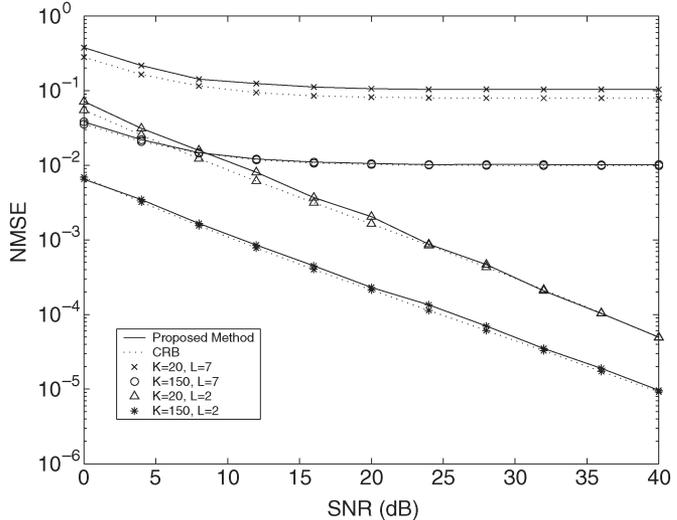


Fig. 6. Comparison of the simulated NMSE and the CRB versus SNR for different channel orders when $n_t = 5$ and constant training sequences are used. The results are plotted for two different values of K , namely 20 and 150.

in the situation where the channel order is small and the SNR is high. In that case, the closed-form estimate is at a significant distance from the CRB. This difference between the CRB and the experimental results originates from the approximations that were needed to derive the approximate closed-form ML channel estimate. These approximations do not hold when the SNR is large, K is small, and the channel order is small. However, when the iterative method is used, the channel estimate converges to \mathbf{h}_{ML} . The experiments presented in Figs. 4 and 5 show that the gap between the CRB and the closed-form estimate is closed after a few iterations. Hence, for small K and high SNR, performing a few iterations allows us to effectively achieve the CRB.

3) *Constant Training Sequences:* In Figs. 6 and 7, we perform a similar analysis for the closed-form ML channel estimate in the context of a constant training sequence. The figures

show us that there is no significant difference between the CRB and the experimental results, except for very small values of K . This improved match between the CRB and the experimental results originates in the fact that we did not need to make any approximation when deriving the expression of the closed-form ML channel estimate in this case. Note that there is no point in using the iterative method in this context, since the closed-form channel estimate corresponds to its convergence point, which is confirmed by experimental results (not shown in the figures).

B. Comparison With Existing Methods

We compare the proposed method with two other methods: a classical training-based ML approach and a more advanced semiblind method based on the EM algorithm presented in [14].

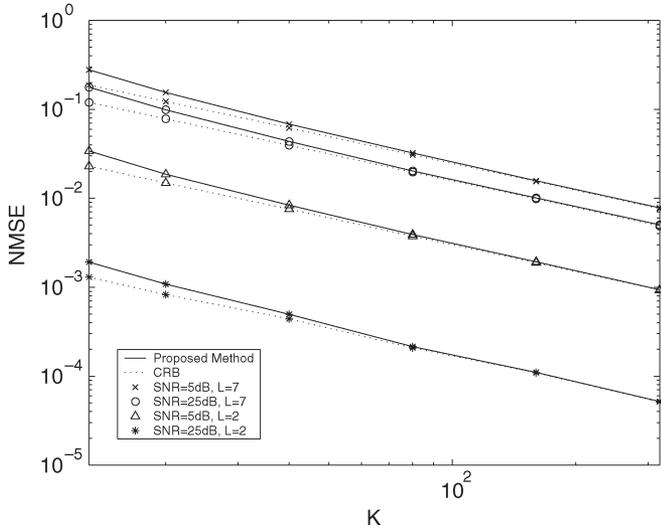


Fig. 7. Comparison of the simulated NMSE and the CRB versus K for different channel orders when $n_t = 5$ and constant training sequences are used. The results are plotted for two different values of the SNR, namely 5 and 25 dB.

Classical training-based ML channel-estimation techniques solely rely on the part of the received symbols that only contain contributions from the known training symbols. They simply discard the received samples that are corrupted by contributions from the unknown data symbols. Such symbols can be observed at the receiver only when $n_t \geq L + 1$. In that case, one can define $\mathbf{T}'_k = \mathbf{T}_k(L + 1 : n_t, :)$ and $\mathbf{u}'_k = \mathbf{u}_k(L + 1 : n_t)$. The ML channel model is then known to be the LS fit of $\mathbf{T}'_k \mathbf{h}$ to \mathbf{u}'_k

$$\mathbf{h}'_{\text{ML}} = \left(\sum_{k=1}^K \mathbf{T}'_k{}^H \mathbf{T}'_k \right)^{-1} \sum_{k=1}^K \mathbf{T}'_k \mathbf{u}'_k.$$

When constant training sequences are used, the solution becomes $\mathbf{h}'_{\text{ML}} = (\sum_{k=1}^K \mathbf{T}'_k{}^H \mathbf{T}'_k)^{-1} \sum_{k=1}^K \mathbf{T}'_k \mathbf{u}'_k$, and only exists when $n_t \geq 2L + 1$. Note that the conditions on the training-sequence length n_t are much more stringent than in the proposed method.

We compare our algorithm with an adapted version of [14]. We adapt that method in order to estimate the channel relying on the channel-output samples contained in the set of \mathbf{u}_k vectors, as exploiting all the channel output samples would yield an untractable complexity. Moreover, as the method cannot be readily adapted to jointly consider all \mathbf{u}_k 's, we compute a channel estimate for each received vector. Considering that the resulting modeling error has a Gaussian distribution, the final ML channel estimate is obtained after computing the mean of the K available channel estimates. Note that since we consider SISO channels rather than the original single-input multiple-output (SIMO) setup, the training sequences must have a minimal length of $n_t \geq L + 1$. If that condition is not fulfilled, the method cannot be applied because of rank-deficiency problems. Hence, it will not be possible to use this method when the channel order gets too large.

We can now compare the results obtained using these methods with the proposed Gaussian ML estimates. The NMSE of

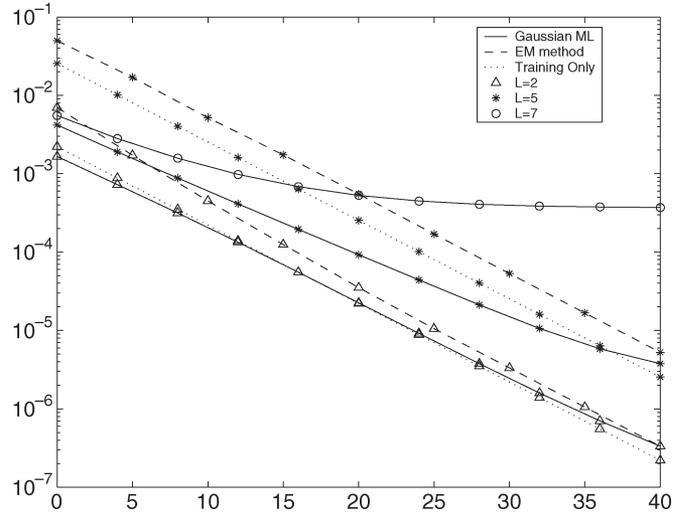


Fig. 8. Simulated NMSE versus SNR for the proposed Gaussian ML method and existing channel-estimation techniques for different channel orders when changing training sequences of length $n_t = 6$ are used and the number of observed blocks is set to $K = 100$.

the different methods is presented as a function of the SNR in Fig. 8. We consider changing training sequences and compare the results for different channel orders when the length of the training sequences is set to $n_t = 6$.

The proposed method outperforms existing ones in all situations, except for short channels and at high SNRs where other methods perform slightly better. We know however that increasing the number of observed blocks K or performing a few iterations would restore the advantage of our Gaussian ML method. When the channel order increases, the advantage of the new method increases, especially at low SNR. When the channel order increases and $n_t < L$, the new method still provides reliable channel estimates while traditional methods cannot be applied anymore. Finally, note that the proposed Gaussian ML method has a complexity that is comparable to those of the training-based method, while the semiblind EM-based method has a significantly higher complexity (up to 200 iterations are needed to reach convergence).

IX. CONCLUSION

In this paper, we presented a new training-based ML channel-identification method where the training sequences can be shorter than the channel impulse-response length. We analyzed two situations: the situation where the same training sequence is repeated at the end of each data block (constant-training-sequence case) and the situation where this training sequence is changed at the end of each data block (changing-training-sequence case). We first proposed an iterative ML method and then derived (approximate) closed-form expressions for the ML channel estimates. The proposed closed-form expressions have low complexity and effectively achieve the CRB in most practical situations. In the few situations where the CRB is not achieved (i.e., low channel order, large SNR, and small number of training sequences), the iterative method can be used and will achieve the CRB in a couple of iterations. The proposed method can be used in white as well as in colored noise conditions.

It clearly outperforms existing training-based and semiblind ML methods.

APPENDIX

In this appendix, we prove that

$$\lim_{K \rightarrow \infty} \hat{\mathbf{Q}}' = \mathbf{Q}.$$

Let us first note that \mathbf{e}_k can be rewritten as

$$\begin{aligned} \mathbf{e}_k &= \mathbf{u}_k - \mathbf{T}_k \left(\sum_{l=1}^K \mathbf{T}_l^H \mathbf{T}_l \right)^{-1} \sum_{l=1}^K \mathbf{T}_l^H \mathbf{u}_l \\ &= \mathbf{T}_k \mathbf{h} + \epsilon_k - \mathbf{T}_k \left(\sum_{l=1}^K \mathbf{T}_l^H \mathbf{T}_l \right)^{-1} \left(\sum_{l=1}^K \mathbf{T}_l^H \mathbf{T}_l \mathbf{h} + \epsilon_l \right) \\ &= \epsilon_k - \mathbf{T}_k \left(\sum_{l=1}^K \mathbf{T}_l^H \mathbf{T}_l \right)^{-1} \sum_{l=1}^K \mathbf{T}_l^H \epsilon_l. \end{aligned}$$

Based on this observation, it is possible to check that the shorthand notation $\hat{\mathbf{Q}}'$ defined in (19) converges to the true \mathbf{Q} when K tends to infinity

$$\begin{aligned} \lim_{K \rightarrow \infty} \hat{\mathbf{Q}}' &= \lim_{K \rightarrow \infty} K^{-1} \sum_{k=1}^K \mathbf{e}_k \mathbf{e}_k^H \\ &= \lim_{K \rightarrow \infty} K^{-1} \sum_{k=1}^K \epsilon_k \epsilon_k^H \\ &\quad - K^{-1} \sum_{k=1}^K \epsilon_k \left(\sum_{l=1}^K \epsilon_l^H \mathbf{T}_l \right) \left(\sum_{l=1}^K \mathbf{T}_l^H \mathbf{T}_l \right)^{-1} \mathbf{T}_k^H \\ &\quad - K^{-1} \sum_{k=1}^K \mathbf{T}_k \left(\sum_{l=1}^K \mathbf{T}_l^H \mathbf{T}_l \right)^{-1} \left(\sum_{l=1}^K \mathbf{T}_l^H \epsilon_l \right) \epsilon_k^H \\ &\quad + K^{-1} \sum_{k=1}^K \mathbf{T}_k \left(\sum_{l=1}^K \mathbf{T}_l^H \mathbf{T}_l \right)^{-1} \left(\sum_{l=1}^K \mathbf{T}_l^H \epsilon_l \right) \\ &\quad \times \left(\sum_{l=1}^K \epsilon_l^H \mathbf{T}_l \right) \left(\sum_{l=1}^K \mathbf{T}_l^H \mathbf{T}_l \right)^{-1} \mathbf{T}_k^H. \end{aligned}$$

Using the central-limit theorem, we can replace the limit of $K^{-1} \sum_k$ by the expected value over k , which yields

$$\begin{aligned} \lim_{K \rightarrow \infty} \hat{\mathbf{Q}}' &= \mathbf{Q} - \mathbf{Q} \mathbf{E} \left\{ \mathbf{T}_k \lim_{K \rightarrow \infty} \left(\sum_{l=1}^K \mathbf{T}_l^H \mathbf{T}_l \right)^{-1} \mathbf{T}_k^H \right\} \\ &\quad - \mathbf{E} \left\{ \mathbf{T}_k \lim_{K \rightarrow \infty} \left(\sum_{l=1}^K \mathbf{T}_l^H \mathbf{T}_l \right)^{-1} \mathbf{T}_k^H \right\} \mathbf{Q}^H \\ &\quad + \mathbf{E} \left\{ \mathbf{T}_k \lim_{K \rightarrow \infty} \left(\sum_{l=1}^K \mathbf{T}_l^H \mathbf{T}_l \right)^{-1} \left(\sum_{l=1}^K \mathbf{T}_l \mathbf{Q} \mathbf{T}_l^H \right)^{-1} \right. \\ &\quad \left. \times \left(\sum_{l=1}^K \mathbf{T}_l^H \mathbf{T}_l \right)^{-1} \mathbf{T}_k^H \right\}. \end{aligned}$$

Since the training sequences have nonzero energy, the limits present in this last expression are equal to zero. Hence, given the finite norm of \mathbf{Q} , all the terms containing these limits are equal to zero, which concludes our proof.

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