Multi-dimensional state smoothing in the presence of non-linear interference

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Component-by-component state smoothing is discussed for multi-dimensional dynamic systems with non-linear random interference such as jamming. Each component of the observation model is a non-linear function of only one state component, arbitrary random interference, and observation noise. Each state component is first approximated by a finite state machine and then, using the Viterbi decoding algorithm of information theory, the state components are sequentially smoothed in parallel. This results in a memory reduction for the implementation of the state smoothing. Simulation results have shown that the proposed scheme performs well, whereas the classical estimation schemes cannot be used, in general, to estimate the states of dynamic models with arbitrary random interference.

1. Introduction

Numerous papers have been written on optimum and sub-optimum recursive state estimation and on their applications since the original work of Kalman and Bucy in the early Sixties (Kalman 1960, Kalman and Bucy 1961). Recursive estimation techniques proposed in these papers have, in general, been developed for state models with linear white disturbance noise and observation models with additive white observation noise (Sage and Melsa 1971, Kailath 1968, 1974, Makhoul 1975, Medich 1973). In addition, several researchers have considered observation models with both additive white observation noise and Markov chains (Nahi 1969, Monzingo 1975). Markov chains are included for considering observation uncertainties. The considered models are either linear or non-linear functions of states. Optimum estimation schemes have been presented for linear models with white gaussian noise. However, sub-optimum estimation schemes have been proposed for non-linear models. These sub-optimum schemes have, in general, been obtained by linearizing non-linear models with a Taylor series expansion. Thus, in some cases, sub-optimum state estimates may diverge from the actual state values because of linearization errors (Miller and Leskiw 1982). In addition, both optimum and sub-optimum estimation schemes treat models that are linear functions of disturbance noise and (additive) observation noise. Hence, these estimation schemes cannot be used, in general, to estimate the states of models with arbitrary random interference in the observation model. If they were used with a zero interference assumption then the resulting state estimates could diverge from the actual state values.

State estimation was also considered by Demirbaş (1984) and Demirbaş and Leondes (1985, 1986) for more general state and observation models, with or without arbitrary random interference. State models are any defined functions (linear or non-
linear) of the state and disturbance noise. Observation models are any defined functions of the state, observation noise, and interference. Furthermore, noises and interference are assumed to be independent from time to time. The proposed estimation schemes estimate the state vector as a vector. As a result, the implementation of these estimation schemes requires an exponentially increasing memory with the dimension of the state vector. Recently, Demirbaş (1989) has proposed a component-by-component state estimation scheme for dynamic models without interference. The implementation of this proposed scheme requires a memory increasing linearly with the dimension of the state vector.

In this paper the state estimation scheme given by Demirbaş (1989) is extended to the state smoothing of dynamic models with non-linear interference. Moreover, the Gallager-type performance of component-by-component state smoothing is discussed for dynamic models with non-linear random interference. The implementation of the proposed smoothing scheme requires a linearly increasing memory, whereas the scheme presented by Demirbaş and Leondes (1986) requires an exponentially increasing memory with the dimension of the state vector.

2. Problem statement

This paper considers the discrete models whose ith state and observation components are defined by

\[ x_i(k + 1) = f_i[k, x(k), w(k)] \]  
\[ z_i(k) = g_i[k, x_i(k), I(k), u(k)] \]

where \( k \) denotes a discrete moment in time; \( w(k) \) is a \( p \times 1 \) disturbance noise vector at time \( k \) with zero mean and known statistics; \( x(0) \) is an \( n \times 1 \) random initial state vector with known statistics, whose ith component is denoted by \( x_i(0) \), where the subscript \( i \) indicates the component label; \( x(k), \ k > 0, \) is an \( n \times 1 \) state vector at time \( k, \) whose ith component is denoted by \( x_i(k) \); \( u(k) \) is an \( I \times 1 \) observation noise vector at time \( k \) with zero mean and known statistics; \( I(k) \) is an \( r \times 1 \) interference vector at time \( k \) with known statistics; \( z(k) \) is an \( n \times 1 \) observation vector at time \( k \) with its ith component, denoted by \( z_i(k) \), is a (linear or non-linear) function of the time \( k \), observation noise vector \( u(k) \), interference vector \( I(k) \) and ith state component only; \( f_i[k, x(k), w(k)] \), and \( g_i[k, x_i(k), I(k), u(k)] \) are either linear or non-linear functions that define the ith state component at time \( k + 1 \) and the observation component at time \( k \) in terms of the state, disturbance noise, observation noise, and interference at time \( k \). Moreover, \( x(0), w(j), \ w(k), I(n), I(p), u(l), \) and \( u(m) \) are assumed to be independent for all \( j, k, n, p, l, \) and \( m \).

The objective is to smooth (estimate) the state sequence \( X^L = \{x(0), x(1), ..., x(L)\} \) using the observation sequence \( Z^L = \{z(1), z(2), ..., z(L)\} \).

State smoothing of the models (1) and (2) has many applications. One of these is target tracking under jamming. In target tracking under jamming, a radar is used for state component measurements. Hence, each observation component (e.g. consider the range measurement) is a function of only one state component (the range), observation noise, and jamming, as in the model of (2), where interference represents the jamming.

3. Smoothing scheme

State smoothing is carried out sequentially, component-by-component, and in parallel. Each state component is first approximated using other state component
estimates and then quantized. This results in an approximation of the state component by a finite state model. This finite state model (or machine) is represented by a trellis diagram, called the component trellis diagram. Then, the component smoothing is carried out by treating the smoothing as a multiple composite hypothesis testing (Van Trees 1968) and using the Viterbi decoding algorithm (Forney 1974, Viterbi and Omura 1979).

Consider a vector \( a(i, k) \) whose \( j \)th component is the estimate of the \( j \)th state component at time \( k \), except for the \( i \)th state component, which is the quantized \( i \)th state component at time \( k \), i.e.

\[
a(j, k) = \begin{cases} \hat{x}_j(k), & \text{if } j \neq i \\ x_{qi}(k), & \text{if } j = i \end{cases}
\]

where \( \hat{x}_j(0) \) is the mean value of the \( j \)th initial state component; \( \hat{x}_j(k) \) is the estimate of the \( j \)th state component at time \( k \), given the observation sequence \( Z^k \); \( x_{qi}(0) \) is the \( i \)th discrete initial state component, which approximates the \( i \)th state component with \( n_i \) possible values denoted by \( x_{qi1}(0), x_{qi2}(0), \ldots \) and \( x_{qin_i}(0) \) (Demirbaş 1984), and these possible values are referred to as the quantization levels of the \( i \)th state component at time zero (or the initial quantization levels of the \( i \)th state component); and \( x_{qi}(k), k > 0, \) is the \( i \)th quantized state component at time \( k \) that is defined by the following finite state model (or machine), which is an approximation of the model (1):

\[
x_{qi}(k + 1) = Q[\hat{x}_i(k), x(k) = a(i, k), w(k) = w_q(k)]
\]

where the possible values of the \( i \)th quantized state component \( x_{qi}(k) \) are denoted by \( x_{qi1}(k), x_{qi2}(k), \ldots, \) and \( x_{qin}(k) \), which are said to be the quantization levels of the \( i \)th state component at time \( k \); \( Q[ \cdot ] \) is the quantizer defined by Demirbaş (1989), which divides the entire real line into non-overlapping intervals (sometimes called gates) of equal length, and which then assigns the centre of each interval to the interval (its length is called the gate size), and \( w_q(k) \) is a discrete disturbance noise vector with \( n_q \) possible values, which approximates the disturbance noise vector \( w(k) \) and these possible values are denoted by \( w_{q1}(k), w_{q2}(k), \ldots, \) and \( w_{qn}(k) \). The observation model of (2) is also approximated by the model

\[
z_i(k) = g_i[k, x_i(k) = x_{qi}(k), I(k) = I_q(k), v(k)]
\]

where \( I_q(k) \) is a discrete interference vector with \( r_q \) possible values which approximates the interference vector \( I(k) \), and these possible values are denoted by \( I_{q1}(k), I_{q2}(k), \ldots, \) and \( I_{qn}(k) \).

The numbers of possible values of discrete random variables and vectors in (3) and (4) are pre-selected, depending upon the desired estimation accuracy with the available memory for the smoothing implementation. The models of (1) and (2) are better approximated by the models of (3) and (4) for larger values of these numbers or smaller gate sizes used, since a random variable or vector is better approximated by a discrete random variable or vector having a greater number of possible values, and since smaller gate sizes result in smaller quantization errors. The number of quantization levels at time \( k \) is determined by the numbers of possible values of discrete random variables and vectors, and the gate sizes used in (3) and (4). The maximum number of quantization levels increases exponentially with time, which can make the smoothing implementation complex. Hence, a compromise between the complexity and estimation accuracy must be made for the smoothing implementation. This determines the pre-selected numbers and gate sizes used.
The finite state model of (3) from time zero to time \( L \) can be represented by a trellis diagram (Demirbaş 1984), referred to as the trellis diagram of the \( i \)th state component—see Fig. 1, whose columns denote possible quantization levels (represented by nodes) which can be taken by the \( i \)th state component from time zero to time \( L \). The directed lines show the possible transitions between the quantization levels. The transition probability \( \pi_{i,m}^n \) from a quantization level \( x_{qi}(k-1) \) (at time \( k-1 \)) to a quantization level \( x_{qi}(k) \) (at time \( k \)) is the probability that the \( i \)th quantized state component takes \( x_{qi}(k) \) at time \( k \) when it was equal to \( x_{qi}(k-1) \) at time \( k-1 \), i.e.

\[
\pi_{i,m}^n = \text{Prob} \{ x_{qi}(k) = x_{qi}(k) \mid x_{qi}(k-1) = x_{qi}(k-1) \}
\]

The trellis diagram for the \( i \)th state component shows a path structure. The quantization levels along the paths of this trellis diagram can be taken by the \( i \)th state component from time zero to time \( L \). Hence, the \( i \)th state component smoothing finds a path through the trellis diagram so that the quantization levels along this path are the smoothed values of the \( i \)th state component from time zero to time \( L \). Finding a path is a multiple composite hypothesis testing problem. The optimum testing rule that minimizes the overall error probability can be stated (Demirbaş 1984) as

\[
\text{choose } H_i \text{ if } M_i^j > M_i^m \text{ for all } m \neq i \tag{5}
\]

where \( H_i \) is the \( i \)th path through the \( i \)th state component trellis diagram, and \( M_i^m \) is, by definition, the metric of the \( m \)th path through the \( i \)th state component trellis diagram, which is defined by

\[
M_i^m = M(x_{qi}(0)) + \sum_{k=1}^{L} M[x_{qi}(k-1) \rightarrow x_{qi}(k)]
\]

where \( x_{qi}(k) \) is the quantization level (or node) at time \( k \) along the \( m \)th path (denoted
by $H_{in}$) of the $i$th state component trellis diagram,

$$M(x_{qi}(0)) \triangleq \ln \{ \text{Prob} \{ x_{qi}(0) = x_{qi}^*(0) \} \}$$

which is, by definition, the metric of the initial node (or quantization level) $x_{qi}^*(0)$, where $\ln$ indicates the natural logarithm.

$$M[x_{qi}(k-1) \to x_{qi}^*(k)] \triangleq \ln \{ \pi_{in} p[z_i(k) \mid x_{qi}(k) = x_{qi}^*(k)] \}$$

which is, by definition, the metric of the branch connecting the node $x_{qi}^*(k-1)$ to the node $x_{qi}^*(k)$, and

$$p(z_i(k) \mid x_{qi}(k) = x_{qi}^*(k)) = \sum_{l=1}^{n_{in}} p(z_i(k) \mid x_{qi}(k) = x_{qi}^*(k), l_d(k) = l_{in}(k)) \times \text{Prob} \{ l_{in}(k) = l_{in}(k) \}$$

which is the conditional density function of the $i$th observation component, given that the $i$th quantized state component $x_{qi}(k)$ is equal to $x_{qi}^*(k)$, where $p[z_i(k) \mid x_{qi}(k) = x_{qi}^*(k), l_d(k) = l_{in}(k)]$ is the conditional density function of $z_i(k)$, given that $x_{qi}(k) = x_{qi}^*(k)$ and $l_d(k) = l_{in}(k)$, and $l_{in}(k)$ is the $i$th possible value of $l_d(k)$.

The optimum decision rule of (5) states that the quantization levels along the path with the greatest metric are the smoothed values of the $i$th state component from time zero to time $L$. For a given observation sequence $Z^k$, if the inequality in (5) becomes an equality for more than one path, the quantization levels along any one of these paths can be chosen as the smoothed values of the $i$th state component from time zero to time $L$. This does not change the overall error probability. Hence, the smoothing of the $i$th state component finds the path with the greatest metric through the $i$th state component trellis diagram. The quantization levels along this path are the smoothed values of the $i$th state component. The path with the greatest metric can be found by the Viterbi decoding algorithm (Demirbaş 1984). It follows from (3) that the $i$th state component quantization levels at time $k+1$ are a function of the other state component estimates $\hat{x}_1(k), \ldots, \hat{x}_{i-1}(k), \hat{x}_{i+1}(k), \ldots, \hat{x}_n(k)$. Thus, these estimates are sequentially obtained by performing the smoothing of the $n$ state components in parallel — see Fig. 2, where VDA represents the Viterbi decoding algorithm. The recursive steps for the state smoothing can be stated as follows:

**Step k (k = 1, 2, ..., L)**

Use the observation sequence $Z^k$ and the VDA for each state component to estimate $n$ state components from time zero to time $k$. This yields the smoothed values $\{ \hat{x}_i(0), \ldots, \hat{x}_i(k) : i = 1, 2, \ldots, n \}$ of the state components from time zero to time $k$, given the observation sequence $Z^k$. The smoothed values of the state components at time $k$ $\{ \hat{x}_1(k), \ldots, \hat{x}_n(k) \}$ are used to find the quantization levels at time $k+1$. This process at Step $L$ yields the smoothed values of the $n$ state components from time zero to time $L$.

At every step, a different smoothed value of a state component at a given time can be obtained, since the path with the greatest metric can change whenever the trellis diagram is extended (Demirbaş 1984).

Assume that, without loss of generality, the numbers of possible values of the discrete initial state components and the discrete disturbance noise vector are component and time-invariant — i.e., $n_{i0} \triangleq n_0$, for $i = 1, 2, \ldots, n$; and $n_k \triangleq n$ for all $k$. It
can be shown (Demirbaş 1989) that the maximum number of quantization levels required for the implementation of the proposed smoothing scheme is \( n_{nomp}' \), whereas the number of maximum quantization levels for the implementation of the scheme presented by Demirbaş and Leondes (1986) is \( n_{nomp} \), where \( p \) and \( n \) are the dimensions of the disturbance noise and state vectors, respectively. Hence, this maximum number is a linear function of the dimension of the state vector for the proposed smoothing scheme, whereas it is an exponential function of the dimension of the state vector for the scheme presented by Demirbaş and Leondes (1986).

\[ \{ \hat{x}_1(0), \hat{x}_2(0), \ldots, \hat{x}_n(0) \} \rightarrow \{ \hat{x}_1(1), \hat{x}_2(1), \ldots, \hat{x}_n(1) \} \rightarrow \{ \hat{x}_1(2), \hat{x}_2(2), \ldots, \hat{x}_n(2) \} \rightarrow \ldots \]

\[ \{ \hat{x}_1(L), \hat{x}_2(L), \ldots, \hat{x}_n(L) \} \]

Figure 2. State smoothing, component by component, in parallel.

4. Performance

The performance of the proposed smoothing scheme depends upon the performances of VDAs used in parallel to smooth the state components. Consider the \( i \)th state component smoothing with the VDA. The performance of this VDA can be quantified by a Gallager-type ensemble upper bound (Gallager 1965, Viterbi and Omura 1979), since the evaluation of the exact error probability or error probability bound for choosing the correct path is complex. It can be shown (Demirbaş 1984) that such an ensemble bound is given by

\[
P_{\text{IE}} \leq F_i(\delta) \prod_{k=1}^{L} \left\{ \sum_{z_{i}(k) \in X_i} q_i(x) p(z_{i}(k) | x)^{1+\delta} \right\}^{1+\delta} dz_{i}(k) \text{ for any } \delta \in [0, 1]
\]

with

\[
F_i(\delta) \triangleq (M_i - 1)^{\delta} \left[ \prod_{k=0}^{L} \left( \frac{q_{\text{max}}}{q_{\text{min}}} \right) ^{b/1+\delta} \right]
\]

where \( P_{\text{IE}} \) is the ensemble-averaged overall error probability for the \( i \)th state component smoothing; \( X_i \) is the set of all possible quantization levels of the \( i \)th state component from time 1 to time \( L \); \( q_i(\cdot) \) is an arbitrary probability density function on
$p(z_i(k)|x)$ is the conditional density function of the $i$th observation component, given that the $i$th state component is equal to $x$; $M_i$ is the number of possible paths through the trellis diagram of the $i$th state component; $\pi_{i0}^{\min}$ and $\pi_{i0}^{\max}$ are the minimum and maximum values of the occurrence probabilities of the possible values of the $i$th discrete initial state component $x_{i0}$; and $\pi_{ik}^{\min}$ and $\pi_{ik}^{\max}$, $k > 0$, are the minimum and maximum values of the transition probabilities from time $k-1$ to time $k$ (in the trellis diagram of the $i$th state component), respectively.

The uniformly weighted ensemble bound with $\delta = 1$ — as the performance measure of the proposed smoothing scheme for the $i$th state component — is used since it is the easiest bound to evaluate (Demirbaş 1984). 'Uniformly weighted' means that $q_i(x) = 1/N_i^\delta$, where $N_i^\delta$ is the number of elements in $X_i^\delta$.

Consider, as an example, the models whose $i$th components are given by

$$x_i(k + 1) = f_i[k, x(k), w(k)] \quad \text{(the state model)} \quad (7)$$

$$z_i(k) = g_i[k, x_i(k), I(k)] + h_i[k, x_i(k), I(k)]v_i(k) \quad \text{(the observation model)} \quad (8)$$

where $x_i(0)$ and $v_i(k)$ are assumed to be gaussian noises with means $m_i$ and 0 and variances $R_{i0}$ and $R_{i0}(k)$, respectively. $h_i[k, x_i(k), I(k)]$ is a given (linear or non-linear) function such that $[h_i(k, x_{q_i}(k), I_{a_i}(k))]^2 R_{i0}(k) \neq 0$, for all $j$, $l$, and $k$. Substituting $p[z_i(k)|x_i(k) = x]$, $\delta = 1$, and $q_i(\cdot) = (1/N_i^\delta)$ into the bound in (6), and using the inequality

$$\left[ \sum_{t} c_t \right]^\eta \leq \sum_{t} [c_t]^\eta \quad \text{for any } c_t \geq 0 \text{ and } \eta \in [0, 1]$$

we obtain the bound

$$\bar{p}_i \leq F_i(1) \left( \frac{1}{N_i^\delta} \right)^2 L \prod_{k=1}^{L} \left\{ \sum_{i} \sum_{j} \sum_{l} \sum_{a} [P(I_{d_i}(k))p(I_{d_j}(k))]^{1/2} L(k, x_1, I_{d_i}(k), x_2, I_{d_j}(k)) \right\} \quad (9)$$

where

$$L[k, x_1, I_1(k), x_2, I_2(k)] \triangleq C_i \exp \left\{ \frac{D_i}{4} \right\}$$

$$C_i \triangleq 2^{1/2}[R_i + R_j]^{1/2}$$

$$D_i \triangleq \frac{[b_{ij}]^2 R_i R_j}{R_i + R_j} - \frac{[g_i(k, x_1, I_1(k))]}{R_i} - \frac{[g_i(k, x_2, I_2(k))]}{R_j}$$

$$b_{ij} \triangleq \frac{[g_i(k, x_1, I_1(k))]}{R_i} + \frac{[g_i(k, x_2, I_2(k))]}{R_j}$$

$$R_i \triangleq [h_i(k, x_1, I_1(k))]^2 R_{i0}(k)$$

$$R_j \triangleq [h_i(k, x_2, I_2(k))]^2 R_{i0}(k)$$

$$P(I_{d_i}(k)) \triangleq \text{Prob} \{I_{d_i}(k) = I_{d_i}(k)\}$$

The bound of (9) is the one used as the performance measure of the proposed scheme for the $i$th state component smoothing of the models (7) and (8).
Figure 3(a). Actual and estimated values of the first state component $x_1(k)$.

Figure 3(b). Error variances and bound for estimates of the first state component $x_1(k)$.
Figure 3(c). Absolute and time-averaged absolute errors for estimates of the first state component $x_1(k)$.  

Figure 3(d). Actual and estimated values of the second state component $x_2(k)$.  

Legend:  
- $\Delta$: Kalman  
- $+$: OOSR
Figure 3 (e). Error variances and bound for estimates of the second state component $x_2(k)$.

Figure 3 (f). Absolute and time-averaged absolute errors for estimates of the second state component $x_2(k)$. 
Figure 4(a). Actual and estimated values of the first state component $x_1(k)$.

Figure 4(b). Error variances and bound for estimates of the first state component $x_1(k)$.
Figure 4(c). Absolute and time-averaged absolute errors for estimates of the first state component $x_1(k)$.

Figure 4(d). Actual and estimated values of the second state component $x_2(k)$. 
Figure 4(e). Error variances and bound for estimates of the second state component $x_2(k)$.

Figure 4(f). Absolute and time-averaged absolute errors for estimates of the second state component $x_2(k)$.
Figure 5 (a). Actual and estimated values of the first state component $x_1(k)$.

Figure 5 (b). Error variances and bound for estimates of the first state component $x_1(k)$. 

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Figure 5(c). Absolute and time-averaged absolute errors for estimates of the first state component $x_1(k)$.

Figure 5(d). Actual and estimated values of the second state component $x_2(k)$.
Figure 5 (e). Error variances and bound for estimates of the second state component $x_2(k)$.

Figure 5 (f). Absolute and time-averaged absolute errors for estimates of the second state component $x_2(k)$. 
5. Simulations

Computer simulations are used to evaluate the performance of the proposed smoothing scheme, and also to observe the divergence of the Kalman filter estimates, assuming zero interference. Many examples are simulated on the IBM 3081K mainframe computer. Examples with white Gaussian noise and interference are considered. Random variables are approximated by the discrete random variables given by Demirbaş (1984). These random variables are also assumed to be time-invariant.

The simulation results of three examples are presented in Figs 3 (a)–5 (f). At the top left-hand corner of each figure, the simulated models and statistics of random variables used are presented. Moreover, in the Figures E(B(·)), VAR (B(·)), and NUM. OF DISC. FOR B(·) denote the mean value and variance of the random variable B(·) and the number of possible values of the discrete random variable used to approximate the random variable B(·); ACTUAL, ODSA, and KALMAN represent the actual values and their estimates by the proposed smoothing scheme and the Kalman filter; AAEOP and AAEK represent the time averaged absolute errors for the proposed smoothing scheme and the Kalman filter; ER. COV. and BOUND indicate the error variances of the Kalman estimates and the performance bound—given by (9)—of the proposed smoothing scheme, respectively. One should note that this bound may become useless (i.e. a number greater than one), depending upon the models and gate sizes used, since some inequalities are used to drive the bound. Even if this bound is less than one, it does not exactly determine the performance of the proposed smoothing scheme, since it is an ensemble bound on the overall error probability (Demirbaş 1984).

The memory requirement for the implementation of the proposed smoothing scheme increases with the numbers of possible values of the discrete random variables and vectors or smaller gate sizes used in the finite state models for state components. Moreover, this memory requirement increases exponentially with time. Hence, by choosing appropriate values for these numbers and gate sizes, one can obtain a desired estimation accuracy, using the available memory, for state smoothing.

The dynamic models, whose simulation results are presented in Figs 3 (a)–5 (f) are non-linear with non-linear interference. In other words, observation component models are non-linear functions of interference. The smoothed values of the state components, using the proposed smoothing scheme, closely follow the actual state component values, while the Kalman estimates assuming zero interference are far from the actual state component values. The proposed smoothing scheme is superior to the Kalman filter, which is incapable of estimating the states of dynamic models with non-linear interference. However, the implementation of the proposed smoothing scheme is more complex than the Kalman filter implementation, since the implementation of the VDA used for each state component smoothing is more complex.

6. Conclusions

The proposed estimation scheme is sub-optimum and applicable to state estimation for multi-dimensional dynamic systems with non-linear noise and interference; whereas the Kalman filter cannot, in general, be used to estimate the states of models with non-linear noise and interference. The implementation of the proposed scheme requires a linearly increasing memory; whereas the implementation of the scheme
presented by Demirbaş and Leondes (1986) requires an exponentially increasing memory with the dimension of the state vector. Moreover, both of these schemes require an exponentially increasing memory with time.

REFERENCES
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