This paper deals with a robust $H_\infty$ deconvolution filtering problem for discrete-time nonlinear stochastic systems with randomly occurring sensor delays. The delayed measurements are assumed to occur in a random way characterized by a random variable sequence following the Bernoulli distribution with time-varying probability. The purpose is to design an $H_\infty$ deconvolution filter such that, for all the admissible randomly occurring sensor delays, nonlinear disturbances, and external noises, the input signal distorted by the transmission channel could be recovered to a specified extent. By utilizing the constructed Lyapunov functional relying on the time-varying probability parameters, the desired sufficient criteria are derived. The proposed $H_\infty$ deconvolution filter parameters include not only the fixed gains obtained by solving a convex optimization problem but also the online measurable time-varying probability. When the time-varying sensor delays occur randomly with a time-varying probability sequence, the proposed gain-scheduled filtering algorithm is very effective. The obtained design algorithm is finally verified in the light of simulation examples.

1. Introduction

Filtering technology is extensively used in many domains, and an important task of filtering is to effectively restore the original signal by removing the distortion from the received signal. The deconvolution method is very useful in filtering field and has been made use of by some kinds of filters, such as self-tuning deconvolution filter and Wiener inverse filter. However, the design process of the deconvolution filter is always difficult because the delivery channels are frequently corrupted by some disturbances, such as nonlinear disturbances, external noises, and sensor delays. Therefore it is very interesting and important to solve the problem encountered in the design procedure of deconvolution filters. In the past decades, considerable attention has been paid to the analysis and design of many kinds of deconvolution filters, and some results have been published. For more details, we refer the readers to [1–3] and the references therein.

As is well known, in some practical fields, such as engineering, biological, medical, and economic systems, and health community, the sensor data is occasionally delayed before they arrive at their respective destinations [4–6]. The occurrence of sensor delay may mainly be caused by the limitations, such as the limited bandwidth of the communication channel, intermittent sensor failure in the measurement, random network congestion, and accidental loss of some collected data in a very noisy environment [6–9]. However, the majority of deconvolution filtering algorithms are based on the measurement outputs without delays. In such case, the traditional filter may fail to work. Hence, in the past decades the filtering problem for the systems with sensor delays has been attracting considerable research interests; see,
for example, [5, 7, 10–12]. Owing to the uncertainty that is widespread in the field of practical engineering, stochastic model has gained more and more attention in many fields, such as physics, economic systems, geomorphology, and gene regulatory networks, and a large number of literatures have been published; see, for example, [4, 13–18].

It should be pointed out that, so far, the Bernoulli distribution has been employed to model some randomly occurring phenomena, such as randomly occurring missing measurements [10] and randomly occurring saturation phenomena, such as physics, economics systems, geomorphology, and gene regulatory networks, and a large number of literatures have been published; see, for example, [4, 13–18]. This model with time-invariant probability has become an effective model of sensor delays. Nevertheless, in the engineering environment, such as industrial automation, unmanned vehicles, real-time distributed decision-making and multiplexed data communication networks [5], and an asynchronous time-division-multiplexed network [6], the sensor delays often occur in a random way and satisfy a time-varying probability distribution. At the same time, the classical filters with fixed gains cannot adapt to the actual time-varying probability distribution. Therefore, there is an urgent need to develop new classical filters with fixed gains cannot adapt to the actual time-varying probability distribution. At the same time, the sensor delays often occur in a random way and satisfy a time-varying probability distribution. In this regard so far; see, for example, [6, 7]. And yet, up to now, to the best of authors’ knowledge, the gain-scheduled deconvolution filtering is still open for discrete-time stochastic systems with randomly occurring sensor delays. It is, therefore, in this paper, we aim to develop an effective gain-scheduled deconvolution filtering algorithm for discrete-time stochastic systems with randomly occurring sensor delays. It is, therefore, in this paper, we aim to develop an effective gain-scheduled deconvolution filtering algorithm for the discrete-time stochastic systems with ROSDs, which is of both theoretical importance and practical significance.

The main contribution of this paper is mainly triplex: (1) for the randomly occurring sensor delay which is one kind of the information incomplete, we exploit a stochastic variable sequence satisfying time-varying Bernoulli distributions to represent the situation of the delayed measurement; (2) a time-varying Lyapunov functional dependent on the distribution probability has been developed and applied to improve the performance of the $H_{\infty}$ deconvolution filter; (3) a new filtering problem with a gain-scheduling approach is addressed for a class of discrete-time nonlinear stochastic systems with randomly occurring sensor delays. It is worth mentioning that, since the considered system involves the probabilistic sensor delays, the sector-like bounded nonlinearity, and the multiplicative noises, it is comprehensive and reasonable. Thanks to the proposed time-varying $H_{\infty}$ deconvolution filter which is designed by employing the gain-scheduling technique, the proposed filtering algorithm can exactly estimate the original input. On account of this merit, the proposed design scheme is more effective and practical.

The rest of this paper is organized as follows. In Section 2, we construct a gain-scheduled deconvolution filter for a class of discrete-time stochastic systems with randomly occurring sensor delays, in which the desired filter gains contain two parts, the fixed gain and time-varying one which is dependent on the time-varying probability. In Section 3, a sufficient condition is derived to guarantee the exponential stability of the augmented system, and the proposed filter is given. By means of constructing Lyapunov functional, we make an assay of the stability for the augmented systems. With the help of the proposed probability-dependent Lyapunov functional, we derived simultaneously another condition indicating the robust ability of the deconvolution filter. A mathematical technique is used to transform the infinite number of inequalities into a finite form. The filter gains are derived from a gain-scheduling approach by resorting to solve a convex optimization problem. In Section 4, an numerical example is presented to demonstrate the reasonable structure and high reliability of the proposed filter. The last section, Section 5, sums up all the arguments in this paper.

Notation. In this paper, $\mathbb{R}^n$, $\mathbb{R}^{n \times m}$, and $\mathbb{I}$ denote, respectively, the $n$-dimensional Euclidean space and the set of all $n \times m$ real matrices, the set of all positive integers. $| \cdot |$ refers to the Euclidean norm in $\mathbb{R}^n$. $I$ denotes the identity matrix of compatible dimension. The notation $X \geq Y$ (resp., $X > Y$), where $X$ and $Y$ are symmetric matrices, means that $X - Y$ is positive semidefinite (resp., positive definite). For a matrix $M$, $M^T$ and $M^{-1}$ represent its transpose and inverse, respectively. The shorthand $\text{diag}(M_1, M_2, \ldots, M_n)$ denotes a block diagonal matrix with diagonal blocks being the matrices $M_1, M_2, \ldots, M_n$. In symmetric block matrices, the symbol $*$ is used as an ellipsis for terms induced by symmetry. Matrices, if they are not explicitly stated, are assumed to have compatible dimensions.

2. Problem Formulation

In this paper, the considered stochastic deconvolution filtering system structure is shown in Figure 1. In the system, the input signal $u(t)$ is transmitted through the channel $\Sigma_c$:

$$
(\Sigma_c): \quad x_c(k + 1) = A_c x_c(k) + B_c u(k) + N_c f(z(k)) + M_c x_c(k) \omega(k),
$$

$$
y(k) = C_c x_c(k), \quad x_c(0) = x_0,
$$

where $x_c(k) \in \mathbb{R}^n$ is the state, $u(k) \in L_{\infty}[0, \infty)$ is the exogenous input signal, $y(k)$ is the actual output, and $x_0$ is the initial state. $\omega(k)$ is a one-dimensional Gaussian white noise sequence satisfying $\mathbb{E}[\omega(k)] = 0$ and $\mathbb{E}[\omega^2(k)] = \sigma^2$, and $z(k) = Z x_c(k)$. For the convenience, $u(k), y(k),$ and $\omega(k)$ are all assumed as scalars. $A_c, B_c, C_c, M_c, N_c,$ and $Z$ are constant matrices with appropriate dimensions.
Remark 1. The deconvolution filtering problem involves the estimation of the signal inputted to a communication channel where the output measurements are disturbed by the channel noise. The channel can be represented by a dynamical system, such as ship roll stabilization systems. In the ideal condition, the input signal would be sent perfectly through the transmission channel without any external influences; however, in the real-world case, from the point of engineering application, most of the communication channels are of limited capacity and suffered from some uncertainties in transmission process such as large environmental noise, channel congestion, and intermittent changes of the signal intensity, and therefore, a few issues have inevitably emerged, for example, channel-induced time delay, the nonlinearity disturbances, channel fading, and so on. In order to describe the channel model, the lower and upper bounds of the sensor delay phenomenon is utilized. Note that gain-scheduled filter is reasonable and the conservatism of which can be reduced since more information about the sensor delay phenomenon is utilized. The designed filter can be scheduled with the time-varying probability, which is able to adapt to changing circumstances naturally. It can be divided into the following several steps. Firstly, compute the constant gains in (10) in terms of the main results that will be developed in this paper. Secondly, estimate/measure the time-varying probability p(k) by statistical tests in real time.

Remark 2. The sensor delays may randomly occur due to some environment reasons, and a Bernoulli distribution model has been introduced in [6, 7] to describe such random phenomenon. However, the Bernoulli distributions in this literature are assumed to be time invariant, which is apparently conservative to deal with the time-varying cases of randomly occurred sensor delays for time-varying systems. In this paper, we will utilize a stochastic variable sequence in (7) satisfying time-varying Bernoulli distributions to express the randomly intermittent phenomenon of the discussed sensor delays.

In order to recover the source signal u(k), the following deconvolution filter structure is considered in this paper:

\[ \Sigma_f : x_f(k + 1) = A_f x_f(k) + B_f y(k), \]  
\[ \hat{u}(k) = C_f x_f(k) + D_f y(k), \]  

where \( x_f(k) \in \mathbb{R}^n \) is the filter state and the matrices \( A_f, B_f, C_f, \) and \( D_f \) are filter parameters to be determined and have the following forms:

\[ A_f = A_{f0} + p(k) A_{fp}, \quad B_f = B_{f0} + p(k) B_{fp}, \]  
\[ C_f = C_{f0} + p(k) C_{fp}, \quad D_f = D_{f0} + p(k) D_{fp}, \]  

where \( A_{f0}, A_{fp}, B_{f0}, B_{fp}, C_{f0}, C_{fp}, D_{f0}, \) and \( D_{fp} \) are the constant filter gains to be designed and \( p(k) \) is the time-varying probability that can be estimated/measured via statistical tests in real time.

Remark 3. Deconvolution filter is a restoration algorithm to remove a wavelet by utilizing a reverse process of convolution. Comparing with conventional filters, it not only can estimate a signal embedded in noise but also can remove the effect of any distortion in the channel systems. Furthermore, it can deal with unknown boundary problem and spatially varying blurs. It is worth mentioning that, different from many conventional deconvolution filters with only constant parameters, the proposed filter gains in (10) include two kinds of filter gains: the fixed parameters \( A_{f0}, A_{fp}, B_{f0}, B_{fp}, C_{f0}, C_{fp}, D_{f0}, \) and \( D_{fp} \) and the time-varying parameter \( p(k) \). The designed filter can be scheduled with the time-varying probability, which is able to adapt to changing circumstances naturally. It can be divided into the following several steps. Firstly, compute the constant gains \( A_{f0}, A_{fp}, B_{f0}, B_{fp}, C_{f0}, C_{fp}, D_{f0}, \) and \( D_{fp} \) in terms of the main results that will be developed in this paper. Secondly, estimate/measure the time-varying probability \( p(k) \) by statistical tests in real time. Lastly, the filter gains can be derived from (10). Obviously, the gain-scheduled filter is reasonable and the conservatism of which can be reduced since more information about the sensor delay phenomenon is utilized. Note that gain-scheduled technique filtering and control problems have become a hot topic and have been intensively researched in the past decades; see, for example, [19–23].
By setting \( \xi(k) = [x^T(k) x^T_f(k)]^T \) and the signal error as \( e(k) = u(k) - \hat{u}(k) \), the dynamics of the filtering process can be derived from (1)-(6) and (8)-(9) as follows:

\[
\xi(k+1) = A(p(k))\xi(k) + D(p(k))\xi(k-d) + Bu(k) + Nf(z(k)) + MJ\xi(k)\omega(k) + (p(k) - \varpi(k))K(p(k))J\xi(k-d),
\]

(11)

\[
e(k) = -[E_{cdf}(p(k)) + (p(k) - \varpi(k))E_{df}(p(k))]\xi(k) - [p(k)E_{df}(p(k)) + (\varpi(k) - p(k))E_{df}(p(k))]\xi(k-d) + u(k),
\]

(12)

where

\[
A(p(k)) = \begin{bmatrix} A_c & 0 \\ (1-p(k))B_fC_c & A_f \end{bmatrix},
\]

\[
B = \begin{bmatrix} B_c \\ 0 \end{bmatrix},
\]

\[
M = \begin{bmatrix} M_c \\ 0 \end{bmatrix},
\]

\[
D(p(k)) = \begin{bmatrix} 0 \\ p(k)B_fC_f \end{bmatrix}0 \]

(13)

\[
K(p(k)) = \begin{bmatrix} 0 \\ B_fC_f \end{bmatrix},
\]

\[
N = \begin{bmatrix} N_c \\ 0 \end{bmatrix},
\]

\[
E_{cdf}(p(k)) = [(1-p(k))D_fC_c C_f],
\]

\[
E_{df}(p(k)) = [D_fC_f 0],
\]

\[
J = [I \ 0].
\]

**Definition 4.** The augmented filtering system (II) is said to be exponentially mean-square stable if, with \( u(k) \equiv 0 \), there exist constant \( \alpha > 0 \) and \( r \in (0,1) \) such that

\[
E\{\|\xi(k)\|^2\} \leq \alpha^k \sup_{-d \leq i < 0} E\{\|\xi(i)\|^2\}, \ k \in \mathbb{R}^+.
\]

(14)

\[
\begin{bmatrix}
Q_d & -Q(p(k)) & * & * & * & * \\
0 & 0 & 0 & 0 & 0 & 0 \\
FZJ & 0 & 0 & 0 & 0 & 0 \\
\sigma^2SJMJ & 0 & 0 & 0 & 0 & 0 \\
\theta(k)S^T(K(p(k)))J & -\theta(k)S^T(K(p(k)))J & 0 & 0 & 0 & -\theta(k)\bar{\Gamma}(k) \\
S^T(A(p(k)) + NF_1ZJ) & S^T D(p(k)) & 0 & 0 & 0 & -\bar{\Gamma}(k)
\end{bmatrix} < 0
\]

(17)

**Definition 5.** Given a scalar \( \gamma > 0 \), the dynamics of the augmented systems (II)-(12) are said to be stochastically stable with disturbance attenuation level \( \gamma \) if it is exponentially mean-square stable, and under zero initial condition, \( \|e(k)\|_{L_2} < \gamma\|u(k)\|_{L_2} \) holds for all nonzero \( u(k) \in L_2[0,\infty) \), where

\[
\|e(k)\|_{L_2} := \left( \mathbb{E}\left\{\sum_{k=1}^{\infty} |e(k)|^2\right\}\right)^{1/2}.
\]

(15)

This paper aims to design a \( H_{\infty} \) deconvolution filter to recover a input signal \( u(k) \) transmitted through a noised channel (1)-(2) such that the closed-loop systems (II)-(12) are stochastically stable with disturbance attenuation level \( \gamma \).

### 3. Main Results

In the proof procedure of theorems presented in this paper, we will use the following lemma.

**Lemma 6.** (Schur complement, see [24]). Given constant matrices \( \Sigma_1, \Sigma_2, \) and \( \Sigma_3 \) where \( \Sigma_1 = \Sigma_1^T \) and \( 0 < \Sigma_2 = \Sigma_2^T \), then \( \Sigma_1 - \Sigma_3\Sigma_2^T \Sigma_3 \geq 0 \) if and only if

\[
\begin{bmatrix}
\Sigma_1 & \Sigma_3 \\
\Sigma_2 & \Sigma_3
\end{bmatrix} \geq 0 \quad \text{or} \quad \begin{bmatrix}
\Sigma_2 & \Sigma_1 \\
\Sigma_3 & \Sigma_1
\end{bmatrix} \geq 0.
\]

(16)

In the following theorem, a sufficient condition is derived for the augmented filtering dynamics (II)-(12) with \( u(k) \equiv 0 \) to ensure the exponential mean-square stability of the considered dynamics.

**Theorem 7.** Consider the augmented filtering system (II) with \( u(k) \equiv 0 \). If there exist positive-definite matrices \( Q_d > 0 \), \( Q(p(k)) > 0 \), and matrix \( S \) such that the following matrix inequalities hold, where

\[
\bar{\Gamma}(k) = -Q(p(k+1)) + S + S^T,
\]

\[
\theta(k) = p(k)(1 - p(k)),
\]

then the augmented dynamics (II) is exponentially stable in mean square sense.
where \( Q(p(k)) \) is a time-varying positive definite matrix sequence dependent on the time-varying probability \( p(k) \):

\[
\mathbb{E} \{ \Delta V(k) \} = \mathbb{E} \{ \xi^T (k + 1) Q(p(k+1)) \xi(k) \}
\]

(20)

Noting that \( \mathbb{E}[\omega(k) - p(k)] = 0 \), \( \mathbb{E}[\omega(k) - p(k)]^2 = p(k)(1 - p(k)) \), and \( \mathbb{E}[\omega^2(k)] = \sigma^2 \), it can be obtained from (11) that

\[
\mathbb{E} \{ \Delta V(k) \} = \mathbb{E} \{ (A(p(k)) \xi(k) + D(p(k)) \xi(k - d) + Bu(k) + Nf(z(k)))^T Q(p(k+1)) (A(p(k)) \xi(k) + D(p(k)) \xi(k - d) + Bu(k) + Nf(z(k))) + p(k)(1 - p(k)) [K(p(k)) \xi(k) - K(p(k)) \xi(k - d)]^T Q(p(k+1)) [K(p(k)) \xi(k) - K(p(k)) \xi(k - d)] - \xi^T (k) [Q(p(k)) - Q_d] \xi(k) - \xi^T (k - d) Q_d \xi(k - d) \}.
\]

(21)

Besides that, by the condition of (4) and (5), (21) can be rewritten as follows:

\[
\mathbb{E} \{ \Delta V(k) \} \leq \mathbb{E} \{ \tilde{\xi}^T (k) \Pi \tilde{\xi}(k) \},
\]

(23)

where \( \tilde{\xi}(k) = [\xi^T (k) \ \xi^T (k - d) \ f_s^T (z(k))]^T \) and

\[
\Pi = \begin{bmatrix}
\Pi_1 & * & * \\
\Pi_2 & \Pi_3 & * \\
\Pi_4 & \Pi_5 & \Pi_6
\end{bmatrix},
\]

(24)

with

\[
\Pi_1 = [A(p(k)) + NF_1 ZJ]^T Q(p(k+1)) J T
\]

\[
\times \begin{bmatrix}
\Pi_1 & * & * \\
\Pi_2 & \Pi_3 & * \\
\Pi_4 & \Pi_5 & \Pi_6
\end{bmatrix},
\]

\[
\Pi_2 = [A(p(k)) + NF_1 ZJ]^T Q(p(k+1)) M J T
\]

\[
\times [A(p(k)) + NF_1 ZJ] \times [A(p(k)) + NF_1 ZJ]^T
\]

\[
+ \sigma^2 \tilde{\xi}^T (k) J^T M^T
\]

\[
\times \begin{bmatrix}
\Pi_1 & * & * \\
\Pi_2 & \Pi_3 & * \\
\Pi_4 & \Pi_5 & \Pi_6
\end{bmatrix},
\]

\[
\Pi_3 = [A(p(k)) + NF_1 ZJ]^T Q(p(k+1)) K(p(k)) J
\]

\[
- (Q(p(k)) - Q_d),
\]

\[
\Pi_4 = [A(p(k)) + NF_1 ZJ]^T Q(p(k+1)) K(p(k)) J
\]

\[
- (Q(p(k)) - Q_d),
\]

\[
\Pi_5 = [A(p(k)) + NF_1 ZJ]^T Q(p(k+1)) K(p(k)) J
\]

\[
- (Q(p(k)) - Q_d),
\]

\[
\Pi_6 = [A(p(k)) + NF_1 ZJ]^T Q(p(k+1)) K(p(k)) J
\]

\[
- (Q(p(k)) - Q_d),
\]
Π₂ = Dᵀ(p(k))Q(p(k+1))
× [A(p(k)) + NF₁ZJ]
− p(k)(1 − p(k))JᵀKᵀ
× (p(k))Q(p(k+1))K(p(k))J,

Π₃ = Dᵀ(p(k))Q(p(k+1))
× [A(p(k)) + NF₁ZJ + FZJ],

Π₄ = NᵀQ(p(k+1))
× [A(p(k)) + NF₁ZJ + FZJ],

Π₅ = NᵀQ(p(k+1))D(p(k)),

Π₆ = NᵀQ(p(k+1))N - 2I.

In the following, we will show that Π < 0 from (17). From the relationship −Q(p(k+1)) + S + Sᵀ > 0 in (17), we can see that S is nonsingular. Performing congruence transformation diag[I, I, I, σ⁻²Γ(k)] to (17), we have

\[
\begin{bmatrix}
Q_d - Q(p(k)) & 0 & -Q_d & 0 & 0 & 0 \\
FZJ & 0 & -2I & 0 & 0 & 0 \\
MJ & 0 & 0 & 0 & 0 & 0 \\
K(p(k))J & -K(p(k))J & 0 & 0 & 0 & 0 \\
A(p(k)) + NF₁ZJ & D(p(k)) & N & 0 & 0 & 0 \\
\end{bmatrix}
\begin{bmatrix}
σ⁻²Γ(k) \\
\Gamma(k) \\
\end{bmatrix}
\leq 0,
\]

where, \( \Gamma(k) = -S⁻¹Q(p(k+1))S⁻¹ + S⁻¹ + S⁻ᵀ \).

Since \( Q⁻¹(p(k+1)) \) is positive definite, we have

\[
Q⁻¹(p(k+1)) + S⁻ᵀQ(p(k+1))S⁻¹ - S⁻¹ - S⁻ᵀ \]
\[= [S⁻ᵀ - Q⁻¹(p(k+1))] \times Q(p(k+1))S⁻¹ \]
\[= [S⁻ᵀ - Q⁻¹(p(k+1))] \times [S⁻¹ - Q⁻¹(p(k+1))] ≥ 0,
\]

therefore \( -Q⁻¹(p(k+1)) ≤ S⁻ᵀQ(p(k+1))S⁻¹ - S⁻¹ - S⁻ᵀ \).

Then, it follows from inequality (27) that

\[
\begin{bmatrix}
Q_d - Q(p(k)) & 0 & -Q_d & 0 & 0 & 0 \\
FZJ & 0 & -2I & 0 & 0 & 0 \\
MJ & 0 & 0 & 0 & 0 & 0 \\
K(p(k))J & -K(p(k))J & 0 & 0 & 0 & 0 \\
A(p(k)) + NF₁ZJ & D(p(k)) & N & 0 & 0 & 0 \\
\end{bmatrix}
\begin{bmatrix}
σ⁻²Γ(k) \\
\Gamma(k) \\
\end{bmatrix}
\leq 0,
\]

In Theorem 7, the sufficient condition ensuring the exponential stability of augmented filtering dynamics (II) has been obtained. Now, we will consider the \( H_∞ \) performance for this dynamics under the zero initial condition.

**Theorem 8.** If there exist positive-definite matrix \( Q_d > 0 \), \( Q(p(k)) > 0 \), and matrix S such that the following matrix inequalities hold:
where $\theta(k)$ has been defined in (18) and

$$
\bar{\Gamma}(k) = -Q(p(k+1)) + S + S^T,
$$

$$
\Phi_1 = S^T K(p(k)) J,
$$

$$
\Phi_2 = S^T (A(p(k)) + NF_1 Z J)
$$

the dynamics of the augmented systems (11)-(12) are stochastically stable with disturbance attenuation level $\gamma$ under the zero initial condition.

Proof. In order to investigate the $H_{\infty}$ performance of the augmented systems (11)-(12), construct a functional as

$$
V(k) := \xi^T(k) Q(p(k)) \xi(k) + \sum_{s=k-d}^{k-1} \xi^T(s) Q_d \xi(s).
$$

Under the zero initial condition and (22), we have

$$
J(N) = \mathbb{E} \left\{ \sum_{k=0}^{N} e^T(k) e(k) - \gamma^2 u^T(k) u(k) \right\}
$$

$$
\leq \mathbb{E} \left\{ \sum_{k=0}^{N} e^T(k) e(k) - \gamma^2 u^T(k) u(k) + \Delta V_k \right\}
$$

$$
\leq \mathbb{E} \left\{ \sum_{k=0}^{N} \xi^T(k) \Pi \xi(k) \right\},
$$

where $\xi(k) = [\xi^T (k) \xi^T (k-d) f^T_s (z(k)) u^T (k)]^T$ and

$$
\Pi = \begin{bmatrix}
\Pi_1 & * & * & * & * & * & * & * \\
\Pi_2 & * & * & * & * & * & * & * \\
\Pi_3 & * & * & * & * & * & * & * \\
\Pi_4 & * & * & * & * & * & * & * \\
\Pi_5 & * & * & * & * & * & * & * \\
\Pi_6 & * & * & * & * & * & * & * \\
\Pi_7 & * & * & * & * & * & * & * \\
\Pi_8 & * & * & * & * & * & * & * \\
\Pi_9 & * & * & * & * & * & * & * \\
\Pi_{10} & * & * & * & * & * & * & * \\
\end{bmatrix},
$$

with

$$
\Pi_1 = [A(p(k)) + NF_1 Z J]^T \\
\times Q(p(k+1)) \\
\times [A(p(k)) + NF_1 Z J] \\
\times \sigma^2 T M^T Q(p(k+1)) M J \\
\times p(k)(1 - p(k)) J^T K^T \\
\times (p(k)) Q(p(k+1)) \\
\times (p(k)) J - (Q(p(k)) - Q_d) \\
\times E_{cdf}^T(p(k)) E_{cdf}(p(k)) \\
\times (p(k)) E_{df}(p(k)),
$$

$$
\Pi_2 = D(p(k))^T \\
\times Q(p(k+1)) [A(p(k)) + NF_1 Z J] \\
\times (p(k)) D(p(k)) + p(k)(1 - p(k)) J^T K^T \\
\times (p(k)) Q(p(k+1)) \\
\times K(p(k)) J + p(k) E_{df}^T(p(k)) \\
\times (p(k)) E_{df}(p(k)) \\
\times (p(k)) E_{df}(p(k)),
$$

$$
\Pi_3 = D(p(k))^T Q(p(k+1)) \\
\times (p(k)) + p(k)(1 - p(k)) \\
\times [K(p(k)) J - Q_d] \\
\times (p(k)) E_{df}(p(k)) + p(k) E_{df}(p(k)),
$$

$$
\Pi_4 = [p(k) E_{df}(p(k))]^T.
$$
By introducing a new slack matrix \( \Psi \)

\[
\Psi(\mathbf{k}) = \begin{bmatrix} \Psi_1 & \Psi_2 & \Psi_3 & \Psi_4 & \Psi_5 & \Psi_6 & \Psi_7 \end{bmatrix}
\]

Theorem 7, we can see that \( \Pi < 0 \). Subsequently, letting \( N \to \infty \), we have

\[
\|e(k)\|_{E_{\Sigma}} < \gamma \|u(k)\|_2.
\]

From Definition 5 and the previous analysis we will safely come to the conclusion that the augmented systems (11)-(12) are stochastically stable with disturbance attenuation level \( \gamma \) under the zero initial condition.

**Remark 9.** For the sake of the desired stability of the augmented systems (11), we propose a Lyapunov functional including the time-varying probability parameters, which reduces the conservatism of the sufficient condition in Theorems 7 and 8. At the same time, in the sufficient condition, we introduce a slack variable \( S \) to decouple the Lyapunov matrices and the filter parameters, which all contain the time-varying probability parameters. Such a technique can bypass the difficulty encountered in the filter design. In the following theorem, the filter design problem is dealt with.

**Theorem 10.** Consider the augmented systems (11)-(12). Assume that there exist positive-definite matrix sequences \( \Theta(p(k)) > 0 \), \( \Theta_d > 0 \) matrix sequences \( \mathbf{B}_j, \mathbf{B}_f, \mathbf{C}_j, \) and \( \mathbf{D}_f \), nonsingular matrices \( S_{21} \), \( R_2 \), and matrix \( R_1 \) such that the following parameter-dependent LMIs hold:

\[
\begin{bmatrix}
\Theta_d - \Theta(p(k)) & \ast & \ast & \ast & \ast & \ast & \ast & \ast & \ast & \ast \\
0 & -\Theta_d & \ast & \ast & \ast & \ast & \ast & \ast & \ast & \ast \\
FZJ & 0 & -2I & \ast & \ast & \ast & \ast & \ast & \ast & \ast \\
0 & 0 & 0 & \ast & \ast & \ast & \ast & \ast & \ast & \ast \\
\Psi_1 & 0 & 0 & 0 & \ast & \ast & \ast & \ast & \ast & \ast \\
\theta(k) \Psi_2(k) & -\theta(k) \Psi_2(k) & 0 & 0 & \ast & \ast & \ast & \ast & \ast & \ast \\
\Psi_3(k) & p(k) \Psi_4(k) & \Psi_5 & \Psi_4 & \Psi_5 & 0 & 0 & \ast & \ast & \ast \\
-\Psi_5(k) & p(k) \Psi_6(k) & 0 & 1 & 0 & 0 & 0 & \ast & \ast & \ast \\
-\theta(k) \Psi_6(k) & \theta(k) \Psi_6(k) & 0 & 0 & 0 & 0 & 0 & \ast & \ast & \ast
\end{bmatrix} < 0, \tag{37}
\]

where

\[
\Psi_1 = \begin{bmatrix} S_{11}^T M_c & 0 \\
R_1^T M_c & 0 \end{bmatrix}, \quad \Psi_2(k) = \begin{bmatrix} \mathbf{B}_f C_c & 0 \\
\mathbf{B}_f C_c & 0 \end{bmatrix},
\]

\[
\Psi_3(k) = \begin{bmatrix} S_{11}^T N_c + (1 - p(k)) \mathbf{B}_f C_c + S_{11}^T N_c F_1 Z \mathbf{B}_f \\\nR_1^T A_c + (1 - p(k)) \mathbf{B}_f C_c + R_1^T N_c F_1 Z \mathbf{B}_f \end{bmatrix}
\]

\[
\Psi_4(k) = \begin{bmatrix} D_f C_c & 0 \\
D_f C_c & 0 \end{bmatrix}, \quad \Psi_5(k) = \begin{bmatrix} S_{11} & S_{11}^T R_1 + R_1^T S_{11} \\\nR_1^T R_1 & R_1^T R_1 \end{bmatrix},
\]

\[
\Psi_6(k) = -\Theta(p(k + 1)) + \begin{bmatrix} S_{11} & S_{11}^T R_1 + R_1^T S_{11} \\
R_1^T R_1 & R_1^T R_1 \end{bmatrix}, \quad \Psi_7(k) = \begin{bmatrix} (1 - p(k)) D_f C_c \\
(1 - p(k)) D_f C_c \end{bmatrix}.
\]

In this case, there exist nonsingular matrices \( S_{21} \) and \( S_{22} \) such that \( R_1 = S_{21}^T S_{22} S_{21} \), and then the constant gains of the desired filter can be obtained as follows:

\[
A_f = S_{21}^T \mathbf{B}_f S_{21} S_{22}, \quad B_f = S_{21}^T \mathbf{B}_f,
\]

\[
C_f = \mathbf{C}_f S_{21}^T S_{22}.
\]

Then, there exists a desired gain-scheduled filter in the form of (8)-(9) such that the dynamics of the augmented systems (11)-(12) are stochastically stable with disturbance attenuation level \( \gamma \) under the zero initial condition.

**Proof.** Let nonsingular matrix variable \( S \) in (30) be partitioned as \( S = [S_{ij}]_{2 \times 2} \), where \( S_{11}, S_{21}, \) and \( S_{22} \) are nonsingular matrices. Introduce matrices
that (11) and (12) are stochastically stable with disturbance such that

\[ R_1 = S_1^2 S_2^2 S_{11}, \quad R_2 = S_2^T S_2^T S_{21}. \]

By performing congruence transformation \( \text{diag}\{\mathcal{F}^{-1}, \mathcal{F}^{-1}, 1, 1, \mathcal{F}^{-1}, \mathcal{F}^{-1}, 1, 1\} \) to (37), we can see that (37) is equivalent to (30). It can now be concluded from Theorem 8 that (11) and (12) are stochastically stable with disturbance attenuation level \( \gamma \) under the zero initial condition.

Because of the time-varying parameter \( p(k) \), the number of LMIs in Theorem 10 is infinite, and all these LMIs bring enormous difficulties to solve. In the next work, we will focus on the challenge and present an effective method to overcome this difficulty.

**Theorem 11.** Consider the augmented systems (11)-(12). Assume that there exist positive positive-definite matrices \( \mathcal{Q}_d > 0 \), \( \mathcal{Q}_p > 0 \), nonsingular matrices \( S_{11} \) and \( R_2 \), and matrices \( R_1 \), \( \mathcal{A}_{f_0}, \mathcal{A}_{fp}, \mathcal{B}_{f_0}, \mathcal{B}_{fp}, \mathcal{C}_{f_0}, \mathcal{C}_{fp}, \mathcal{D}_{f_0} \), and \( D_{fp} \), such that the following LMIs hold:

\[
\begin{bmatrix}
\mathcal{Q}_d - \mathcal{Q}_d' & * & * & * & * & * & * & * & * \\
0 & -\mathcal{Q}_d & * & * & * & * & * & * & * \\
F \Psi_{ij} & 0 & -2I & * & * & * & * & * & * \\
0 & 0 & 0 & -\gamma^2 & * & * & * & * & * \\
\Psi_{ijr} & 0 & 0 & 0 & -\delta^i \Psi_{6l} & * & * & * & * \\
\Psi_{ijr} & -\Psi_{ijr} & 0 & 0 & 0 & -\theta^i \Psi_{6l} & * & * & * \\
\Psi_{ijr} & \Psi_{ijr} & \Psi_4 & \Psi_5 & 0 & 0 & -\Psi_{6l}^l & * & * \\
-\theta^i \Psi_{8l} & \theta^i \Psi_{8l} & 0 & 0 & 0 & 0 & 0 & 0 & -\theta^i \\
\end{bmatrix} < 0,
\]

(41)

for \( i, j, r, l = 1, 2 \),

\[
\begin{align*}
\Psi_{3r} & = \left[ S^T_{11} A_c + (1-p_j) \mathcal{B}_{f} C_c + S^T_{11} N_c F_1 Z \mathcal{A}_{f} \right], \\
\Psi_{7r} & = \left[ S^T_{11} A_c + (1-p_j) \mathcal{B}_{f} C_c + R^T_{11} N_c F_1 Z \mathcal{A}_{f} \right], \\
\Psi_{6l} & = - \left( \mathcal{Q}_0 + p \mathcal{Q}_p \right) + \left[ S_{11}^T + S_{11}^T \right], \\
\Psi_{ijr} & = \left[ (1-p_j) D_{f}^T C_c \right], \\
\Psi_{r} & = \mathcal{A}_{f_0} + p \mathcal{A}_{fp}, \\
\Psi_{fp} & = \mathcal{A}_{f_0} + p \mathcal{A}_{fp}, \\
\psi_{6l} & = \psi_{0l} + p \psi_{p}, \\
\theta_{ijr} & = \left[ \psi_{ijr} \right], \\
\psi_{8l} & = \left[ D_{f}^T C_c \right],
\end{align*}
\]

(42)

and \( \Psi_1, \Psi_4, \text{and} \Psi_5 \) have been defined in (38).

In this case, there exist nonsingular matrices \( S_{21} \) and \( S_{22} \) such that \( R_2 = S_{21}^T S_{22}^T S_{21} \), and then the constant gains of the desired filter can be obtained as follows:

\[
A_{f_0} = S_{21}^T \mathcal{A}_{f_0} S_{21}^T S_{22}, \\
A_{fp} = S_{21}^T \mathcal{A}_{fp} S_{21}^T S_{22},
\]

(43)

Then, a gain-scheduled filter can be obtained in the form of (8)-(9) such that the dynamics of the augmented systems (11)-(12) are stochastically stable with disturbance attenuation level \( \gamma \) under the zero initial condition.

**Proof.** Firstly, choose the probability-dependent Lyapunov matrices as

\[
\mathcal{Q}(p(k)) = \mathcal{Q}_0 + p(k) \mathcal{Q}_p,
\]

(44)

where \( \mathcal{Q}_0 > 0 \) and \( \mathcal{Q}_p > 0 \). Setting

\[
\lambda_1(k) = \frac{p_2 - p(k)}{p_2 - p_1}, \quad \lambda_2(k) = \frac{p(k) - p_1}{p_2 - p_1},
\]

(45)

we have

\[
\lambda_1(k) + \lambda_2(k) = 1, \quad \lambda_i(k) \geq 0 \quad (i = 1, 2),
\]

(46)

Similarly, letting

\[
\mu_1(k) = \frac{p_2 - p(k + 1)}{p_2 - p_1}, \quad \mu_2(k) = \frac{p(k + 1) - p_1}{p_2 - p_1},
\]

(47)

we have

\[
\mu_1(k) + \mu_2(k) = 1, \quad \mu_i(k) \geq 0 \quad (i = 1, 2),
\]

(48)
Algorithm 13. The gain-scheduled filter design algorithm.

**Step 1.** Given the initial values for the positive integer $N_q$, the initial state $\rho_0$, the constants $p_1$ and $p_2$, and the matrices $A_c$, $B_c$, $C_c$, $M_c$, $N_c$, $F_1$, $F_2$, and $Z$ select appropriate initial state estimate $\hat{\rho}_f$, and set $k = 0$.

**Step 2.** Solve the LMI in (41) to obtain the positive-definite matrices $\mathcal{G}_0$, $\mathcal{G}_p$, and $\mathcal{G}_d$, matrices $\mathcal{U}_0$, $\mathcal{U}_f$, $\mathcal{B}_0$, $\mathcal{B}_f$, $\mathcal{C}_0$, $\mathcal{C}_f$, $R_1$, $R_2$, and $S_{11}$. Choose appropriate nonsingular matrices $S_{21}$ to derive $A_f$, $A_{fp}$, $B_{fp}$, $B_{f}$, $C_f$, and $C_{fp}$.

**Step 3.** Based on the measured time-varying parameter $p(k)$, derive the filter gains $A_f$, $B_f$, $C_f$, and $D_f$ by (10), the state $x_f(k+1)$ by (9), and the estimation of $u(k)$. Then, set $k = k + 1$.

**Step 4.** If $k < N$, then go to Step 3; otherwise go to Step 5.

**Step 5.** Stop.

**Remark 12.** In Theorem II, we convert infinite LMIs to finite ones by turning the time-varying parameter $p(k)$ into the polytopic form. By such a transformation, the constant gains of the desired gain-scheduled filter can be easily derived in terms of the available LMI toolbox by using the computationally appealing gain-scheduled deconvolution filter design algorithm listed as follows.

Algorithm 13. The gain-scheduled filter design algorithm.

**Step 1.** Given the initial values for the positive integer $N_q$, the initial state $\rho_0$, the constants $p_1$ and $p_2$, and the matrices $A_c$, $B_c$, $C_c$, $M_c$, $N_c$, $F_1$, $F_2$, and $Z$ select appropriate initial state estimate $\hat{\rho}_f$, and set $k = 0$.

**Step 2.** Solve the LMI in (41) to obtain the positive-definite matrices $\mathcal{G}_0$, $\mathcal{G}_p$, and $\mathcal{G}_d$, matrices $\mathcal{U}_0$, $\mathcal{U}_f$, $\mathcal{B}_0$, $\mathcal{B}_f$, $\mathcal{C}_0$, $\mathcal{C}_f$, $R_1$, $R_2$, and $S_{11}$. Choose appropriate nonsingular matrices $S_{21}$ to derive $A_f$, $A_{fp}$, $B_{fp}$, $B_{f}$, $C_f$, and $C_{fp}$.

**Step 3.** Based on the measured time-varying parameter $p(k)$, derive the filter gains $A_f$, $B_f$, $C_f$, and $D_f$ by (10), the state $x_f(k+1)$ by (9), and the estimation of $u(k)$. Then, set $k = k + 1$.

**Step 4.** If $k < N$, then go to Step 3; otherwise go to Step 5.

**Step 5.** Stop.

**Remark 14.** In Algorithm 13, detailed steps have been given for the gain-scheduled deconvolution filter design problem according to Theorem II. By employing this algorithm along with the LMI toolbox, the time-varying filter gains can be easily derived from the measured/estimated time-varying missing probability $p(k)$ in real time.

### 4. An Illustrative Example

In this section, an example is given to design the deconvolution filters for stochastic systems with randomly occurring sensor delays.

The system parameters are given as follows:

- $A_c = \begin{bmatrix} 0.601 & -0.065 \\ 0 & 0.420 \end{bmatrix}$, $M_c = \begin{bmatrix} 0.013 & 0 \\ 0 & 0.024 \end{bmatrix}$,
- $N_c = \begin{bmatrix} 0.014 & 0 \\ 0 & 0.062 \end{bmatrix}$, $Z = \begin{bmatrix} 0.291 & 0 \\ 0 & 0.599 \end{bmatrix}$,
- $F_1 = \begin{bmatrix} 0.159 & 0 \\ 0 & 0.311 \end{bmatrix}$, $F_2 = \begin{bmatrix} 0.409 & 0 \\ 0 & 1.501 \end{bmatrix}$, 
- $C_c = \begin{bmatrix} 0.22 & 0.075 \end{bmatrix}$, $B_c = \begin{bmatrix} 0.109 \\ 0.081 \end{bmatrix}$, $p_1 = 0.23$, $p_2 = 0.45$, $\sigma = 1$.

The measurable time-varying probability sequence is assumed as $0.23e^{0.0168k}$. Then, the constant filter parameters $A_{f0}$, $A_{fp}$, $B_{f0}$, $B_{fp}$, $C_{f0}$, $C_{fp}$, $D_{f0}$, and $D_{fp}$ can be obtained as follows:

- $A_{f0} = \begin{bmatrix} 0.3998 & 0.2138 \\ -0.1733 & -0.0501 \end{bmatrix}$, $B_{f0} = \begin{bmatrix} -0.0521 \\ 0.0364 \end{bmatrix}$,
- $C_{f0} = \begin{bmatrix} -0.0266 & -0.0249 \end{bmatrix}$, $D_{f0} = 16$, 
- $A_{fp} = \begin{bmatrix} -0.0109 & -0.0072 \\ -0.0246 & -0.0160 \end{bmatrix}$, $B_{fp} = \begin{bmatrix} -0.0230 \\ 0.0020 \end{bmatrix}$,
- $C_{fp} = \begin{bmatrix} -0.0794 & -0.0520 \end{bmatrix}$, $D_{fp} = 17$.

Figure 2 includes the response curves of input signal $u(k)$ and the simulation results of estimation $\hat{u}(k)$. Figure 3 gives the time-varying missing probability $p(k)$, and the corresponding filter parameters are given in Table 1. The simulation results have illustrated the rationality and effectiveness of the previous theoretical analysis.

### 5. Conclusions

This paper has dealt with the deconvolution filtering problem for a class of discrete-time stochastic systems with randomly occurring sensor delays, nonlinear disturbances, and external stochastic noises. We assume the sensor delays to be randomly occurring, and the occurring way is modeled by a stochastic variable sequence satisfying time-varying Bernoulli distributions. A sufficient condition has been derived to guarantee the stability of the considered stochastic systems by constructing probability-dependent Lyapunov functional and employing convex optimization method. Through some mathematical transformation, we convert the matrix inequalities into solvable form, and then a finite set of inequalities for designing the desired filter are obtained. The proposed gain-scheduled filters include both constant parameters and time-varying gains which can be updated online according to the measurable missing probabilities in real time. The desired filters can be obtained by solving a set of LMIs relying on the time-varying feature of sensor delays. By using the obtained filter, we can accurately estimate the input signal distorted by the noisy transmission channel and the delayed sensor outputs. A simulation example is exploited to illustrate the effectiveness of the proposed design scheme.
Table 1: The time-varying filter parameters.

<table>
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<tr>
<th>k</th>
<th>p(k)</th>
<th>A_f</th>
<th>B_f</th>
<th>C_f</th>
<th>D_f</th>
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<td>1</td>
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<td>0.3972 0.2121</td>
<td>-0.0574</td>
<td>[0.3972 0.2121 0.0369</td>
<td>19.9762</td>
</tr>
<tr>
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<td>0.3972 0.2121</td>
<td>-0.0575</td>
<td>[0.3972 0.2121 0.0369</td>
<td>20.0436</td>
</tr>
<tr>
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<td>0.3971 0.2121</td>
<td>-0.0576</td>
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<td>20.1121</td>
</tr>
<tr>
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<td>-0.0577</td>
<td>[0.3971 0.2120 0.0369</td>
<td>20.1818</td>
</tr>
</tbody>
</table>

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References


