

DetServ: Network Models for Real-Time QoS Provisioning in SDN-based Industrial Environments

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Based on:

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ТЛП



£2.000.000 turbine destroyed

Turbine stopped when wind too strong

60.000 people without electrity

Turbine disconnected from the grid when it fails

£2.000.000 turbine destroyed

Turbine stopped when wind too strong

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Turbine disconnected from the grid when it fails

sensor – controller – actuator communication







Turbine disconnected from the grid when it fails

£2.000.000 turbine destroyed

60.000 people without electrity

In order to avoid such losses and damages

sensor – controller – actuator communication

must be

deterministically delay bounded



deterministically delay bounded





deterministically delay bounded





SoA proprietary technologies are typically costly and not interoperable



performance (resource efficiency)

ТШ





Embed **f** such that

i. its delay t_r is guaranteed

ii. the guarantees provided to previously embedded flows are still valid





Embed **f** such that

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ii. the guarantees provided to previously embedded flows are **still valid**





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Embed **f** such that i. its delay t_r is guaranteed ii the guarantees provided to previously embedded flows are still valid getDelay() embed new flow **f** with delay \mathbf{t}_{f} **Delay-constrained** Network Model Routing registerPath() hasAccess() **OpenFlow rules**







ТШП

The delay of a route depends on

- # the physical links
- # how the flow is scheduled at each node

the physical **links** #

assuming priority scheduling (cheap and ubiquitous)

the **queue** at which the flow is schedule at each node

ПП

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ТЛП

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assuming priority scheduling $\boldsymbol{\zeta}$

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#

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the **queue** at which the flow is schedule at each node

with, e.g., 3 priority queues at each output port





the physical links -

assuming priority scheduling

(cheap and ubiquitous)

the **queue** at which the flow is schedule at each node

QUEUE LINK TOPOLOGY

with, e.g., 3 priority queues at each output port

Performing route selection on this topology defines both





the physical links 🖛

assuming priority scheduling

(cheap and ubiquitous)



QUEUE LINK TOPOLOGY

with, e.g., 3 priority queues at each output port

Performing route selection on this topology defines both



ТШ



As we need deterministic delay guarantees,

deterministic network calculus

is a perfect candidate modeling tool!



 $\begin{array}{ll} R_{u,v} & \mbox{capacity of link } (u,v) \\ p \in [1,Q_{u,v}] & \mbox{queue priorities (1 being highest)} \\ (u,v,p) & \mbox{queue with priority } p \mbox{ at link } (u,v) \\ \mathbf{U}_R[u,v,p] & \mbox{rate of flow through queue } (u,v,p) & \mbox{We as} \\ \mathbf{U}_B[u,v,p] & \mbox{burst of flow through queue } (u,v,p) & \mbox{We as} \\ l_{u,v,p}^{max} & \mbox{maximum packet size through queue } (u,v,p) \end{array}$

We assume **token bucket** flows through each queue

The service curve for priority queue (u, v, p) is given by

assuming priority scheduling

(cheap and ubiquitous)

$$\left(R_{u,v}t - t\sum_{j=1}^{p-1} \mathbf{U}_R[u,v,j] - \sum_{j=1}^{p-1} \mathbf{U}_B[u,v,j] - \max_{p+1 \le j \le Q_{u,v}} \{l_{u,v,j}^{max}\} - l_{u,v,p}^{max}\right)^+$$



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assuming priority scheduling

(cheap and ubiquitous)

The service curve for priority queue (u, v, p) is given by



whole link service

service used by higher priority queues

one packet from a lower priority queue (non-preemptive scheduling)

per packet delay

The service curve for priority queue (u, v, p) is given by

$$\beta_{u,v,p} = \left(R_{u,v}t - t\sum_{j=1}^{p-1} \mathbf{U}_R[u,v,j] - \sum_{j=1}^{p-1} \mathbf{U}_B[u,v,j] - \max_{p+1 \le j \le Q_{u,v}} \{l_{u,v,j}^{max}\} - l_{u,v,p}^{max} \right)^+$$



ПΠ

The path P_f of a flow f must be chosen such that if fulfills the delay requirement t_f of the flow,

$$\sum_{(u,v,p)\in P_f} \mathbf{T}[u,v,p] \le t_f$$

and we must ensure that no buffer overflow occurs

$$B_{max}(u, v, p) \leq \mathbf{A}_B[u, v, p] \qquad \forall \ (u, v, p)$$

Buffer capacity of a queue



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Buffer capacity of a queue

Must be fulfilled **at all times**, for all the flows



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Must be fulfilled **at all times**, for all the flows

We have...

$$\mathbf{T}[u, v, p] = \frac{\sum_{j=1}^{p} \mathbf{U}_{B}[u, v, j] + \max_{p+1 \le j \le Q_{u,v}} \{l_{u,v,j}^{max}\} + l_{u,v,p}^{max}}{R_{u,v} - \sum_{j=1}^{p-1} \mathbf{U}_{R}[u, v, j]}$$

$$B_{max}(u,v,p) = \mathbf{U}_B[u,v,p] + \mathbf{U}_R[u,v,p] \frac{\sum_{j=1}^{p-1} \mathbf{U}_B[u,v,j] + \max_{p+1 \le j \le Q_{u,v}} \{l_{u,v,j}^{max}\} + l_{u,v,p}^{max}}{R_{u,v} - \sum_{j=1}^{p-1} \mathbf{U}_R[u,v,j]}$$



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Buffer capacity of a queue
We have...
Dependence on other flows embedded at the same link
$$\mathbf{T}[u,v,p] = \frac{\sum_{j=1}^{p} \mathbf{U}_{B}[u,v,j] + \max_{p+1 \leq j \leq Q_{u,v}} \left[\lim_{u,v,j} + \lim_{p+1 \leq j \leq Q_{u,v}} \left[\lim_{u,v,j} + \lim_{u,v,j} \left[\lim_{u,v,j} + \lim_{u,v,j} + \lim_{u,v,j} \left[\lim_{u,v,j} + \lim_{u,v,j} + \lim_{u,v,j} \left[\lim_{u,v,j} + \lim_{u,v,j} + \lim_{u,v$$

$$\sum_{(u,v,p)\in P_f} \mathbf{T}[u,v,p] \le t_f \qquad \forall f$$

$$B_{max}(u,v,p) \le \mathbf{A}_B[u,v,p] \qquad \forall \ (u,v,p)$$

Should we re-check all the previously embedded flows?

No, it does not scale!

$$\sum_{(u,v,p)\in P_f} \mathbf{T}[u,v,p] \le t_f \quad \forall f$$
$$B_{max}(u,v,p) \le \mathbf{A}_B[u,v,p] \quad \forall (u,v,p)$$

Should we re-check all the previously embedded flows?

No, it does not scale!

→ Define upper bounds which are independent of the state of the network!



$$\sum_{(u,v,p)\in P_f} \mathbf{T}[u,v,p] \le t_f \quad \forall f$$
$$B_{max}(u,v,p) \le \mathbf{A}_B[u,v,p] \quad \forall \ (u,v,p)$$

Should we re-check all the previously embedded flows?

No, it does not scale!

→ Define upper bounds which are independent of the state of the network!



Let's find an upper bound **independent of the network state**...

$$\mathbf{T}[u, v, p] = \frac{\sum_{j=1}^{p} \mathbf{U}_{B}[u, v, j] + \max_{p+1 \le j \le Q_{u,v}} \{l_{u,v,j}^{max}\} + l_{u,v,p}^{max}}{R_{u,v} - \sum_{j=1}^{p-1} \mathbf{U}_{R}[u, v, j]}$$

$$B_{max}(u,v,p) = \mathbf{U}_B[u,v,p] + \mathbf{U}_R[u,v,p] \frac{\sum_{j=1}^{p-1} \mathbf{U}_B[u,v,j] + \max_{p+1 \le j \le Q_{u,v}} \{l_{u,v,j}^{max}\} + l_{u,v,p}^{max}}{R_{u,v} - \sum_{j=1}^{p-1} \mathbf{U}_R[u,v,j]}$$
ТШ

Packets cannot be bigger than the biggest Ethernet frame size

$$\mathbf{T}[u, v, p] = \frac{\sum_{j=1}^{p} \mathbf{U}_{B}[u, v, j] + 1542 + 1542}{R_{u,v} - \sum_{j=1}^{p-1} \mathbf{U}_{R}[u, v, j]}$$
$$B_{max}(u, v, p) = \mathbf{U}_{B}[u, v, p] + \mathbf{U}_{R}[u, v, p] \frac{\sum_{j=1}^{p-1} \mathbf{U}_{B}[u, v, j] + 1542 + 1542}{R_{u,v} - \sum_{j=1}^{p-1} \mathbf{U}_{R}[u, v, j]}$$



 $\mathbf{A}_{R}[u,v,p]$ defined as the maximum rate that can be accepted at a queue

$$\mathbf{T}[u, v, p] = \frac{\sum_{j=1}^{p} \mathbf{U}_{B}[u, v, j] + 1542 + 1542}{R_{u,v} - \sum_{j=1}^{p-1} \mathbf{A}_{R}[u, v, p]}$$
$$B_{max}(u, v, p) = \mathbf{U}_{B}[u, v, p] + \mathbf{A}_{R}[u, v, p] \frac{\sum_{j=1}^{p-1} \mathbf{U}_{B}[u, v, j] + 1542 + 1542}{R_{u,v} - \sum_{j=1}^{p-1} \mathbf{A}_{R}[u, v, p]}$$





Limit bursts such that no buffer overflow occurs

$$\begin{aligned} \mathbf{T}[u, v, p] &= \frac{\sum_{j=1}^{p} \mathbf{U}_{B}[u, v, j] + 1542 + 1542}{R_{u,v} - \sum_{j=1}^{p-1} \mathbf{A}_{R}[u, v, p]} \\ B_{max}(u, v, p) &= \mathbf{U}_{B}[u, v, p] + \mathbf{A}_{R}[u, v, p] \frac{\sum_{j=1}^{p-1} \mathbf{U}_{B}[u, v, j] + 1542 + 1542}{R_{u,v} - \sum_{j=1}^{p-1} \mathbf{A}_{R}[u, v, p]} \end{aligned}$$



Limit bursts such that no buffer overflow occurs

$$\begin{aligned} \mathbf{T}[u, v, p] &= \frac{\sum_{j=1}^{p} \mathbf{M}_{B}[u, v, p] + 1542 + 1542}{R_{u,v} - \sum_{j=1}^{p-1} \mathbf{A}_{R}[u, v, p]} \\ B_{max}(u, v, p) &= \mathbf{M}_{B}[u, v, p] + \mathbf{A}_{R}[u, v, p] \frac{\sum_{j=1}^{p-1} \mathbf{M}_{B}[u, v, p] + 1542 + 1542}{R_{u,v} - \sum_{j=1}^{p-1} \mathbf{A}_{R}[u, v, p]} \end{aligned}$$

Limit bursts such that no buffer overflow occurs

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The maximum bursts $\mathbf{M}_B[u, v, p]$ can be computed recursively

$$\mathbf{M}_{B}[u, v, 1] + \mathbf{A}_{R}[u, v, 1] \frac{3084}{R_{u,v}} = \mathbf{A}_{B}[u, v, 1]$$

ТЛП

Limit bursts such that no buffer overflow occurs

$$\begin{aligned} \mathbf{T}[u, v, p] &= \frac{\sum_{j=1}^{p} \mathbf{M}_{B}[u, v, p] + 1542 + 1542}{R_{u,v} - \sum_{j=1}^{p-1} \mathbf{A}_{R}[u, v, p]} \\ B_{max}(u, v, p) &= \mathbf{M}_{B}[u, v, p] + \mathbf{A}_{R}[u, v, p] \frac{\sum_{j=1}^{p-1} \mathbf{M}_{B}[u, v, p] + 1542 + 1542}{R_{u,v} - \sum_{j=1}^{p-1} \mathbf{A}_{R}[u, v, p]} \end{aligned}$$

The maximum bursts $\mathbf{M}_B[u,v,p]$ can be computed recursively

$$\mathbf{M}_{B}[u, v, 2] + \mathbf{A}_{R}[u, v, 2] \frac{\mathbf{M}_{B}[u, v, 1] + 3084}{R_{u,v} - \mathbf{A}_{R}[u, v, 1]} = \mathbf{A}_{B}[u, v, 1]$$

etc.

$$\mathbf{T}[u, v, p] \leq \frac{\sum_{j=1}^{p} \mathbf{M}_{B}[u, v, j] + 3084}{R_{u,v} - \sum_{j=1}^{p-1} \mathbf{A}_{R}[u, v, j]} \triangleq \mathbf{T}^{MHM}[u, v, p]$$
defined per queue by a resource allocation algorithm

ПΠ





$$\mathbf{T}[u, v, p] \leq \frac{\sum_{j=1}^{p} \mathbf{M}_{B}[u, v, j] + 3084}{R_{u,v} - \sum_{j=1}^{p-1} \mathbf{A}_{R}[u, v, j]} \triangleq \mathbf{T}^{MHM}[u, v, p]$$

$$\frac{\mathbf{The Multi-Hop Model (MHM)}}{\operatorname{return T}^{MHM}[u, v, p]}$$

$$\operatorname{return T}^{MHM}[u, v, p] \underbrace{\mathsf{getDelay}()}_{\operatorname{registerPath}()} \underbrace{\mathsf{Network}}_{\operatorname{Model}}$$

$$\operatorname{check M}_{B}[u, v, p], \mathbf{A}_{R}[u, v, p] \text{ are not exceeded}}$$





$$\mathbf{T}^{MHM}[u, v, p] = \frac{\sum_{j=1}^{p} \mathbf{M}_{B}[u, v, j] + 3084}{R_{u,v} - \sum_{j=1}^{p-1} \mathbf{A}_{R}[u, v, j]}$$



$$\mathbf{T}^{MHM}[u, v, p] = \frac{\sum_{j=1}^{r} \mathbf{M}_{B}[u, v, j] + 3084}{R_{u,v} - \sum_{j=1}^{p-1} \mathbf{A}_{R}[u, v, j]}$$



$$\mathbf{T}^{MHM}[u, v, p] = \frac{\sum_{j=1}^{p} \mathbf{M}_{B}[u, v, j] + 3084}{R_{u,v} - \sum_{j=1}^{p-1} \mathbf{A}_{R}[u, v, j]}$$



$$\mathbf{T}^{MHM}[u, v, p] = \frac{\sum_{j=1}^{p} \mathbf{M}_{B}[u, v, j] + 3084}{R_{u,v} - \sum_{j=1}^{p-1} \mathbf{A}_{R}[u, v, j]}$$



$$\mathbf{T}^{MHM}[u, v, p] = \frac{\sum_{j=1}^{j} \mathbf{M}_B[u, v, j] + 5064}{R_{u,v} - \sum_{j=1}^{p-1} \mathbf{A}_R[u, v, j]}$$







$$R_{u,v} - \sum_{j=1}^{r} \mathbf{A}_{R}[u, v, j]$$

$$\beta_{u,v,p} = \left(R_{u,v}t - t\sum_{j=1}^{p-1} \mathbf{A}_R[u,v,j] - \sum_{j=1}^{p-1} \mathbf{M}_B[u,v,j] - 3084 \right)$$

$$\mathbf{T}^{MHM}[u, v, p] = \frac{\sum_{j=1}^{p} \mathbf{M}_{B}[u, v, j] + 3084}{R_{u,v} - \sum_{j=1}^{p-1} \mathbf{A}_{R}[u, v, j]}$$



$$\beta = \left(\begin{array}{ccc} p-1 & p-1 \\ p-1 & p-1 \end{array} \right)^{p-1} \mathbf{M} = \begin{bmatrix} a & a & i \end{bmatrix} = 2084 \right)^{+}$$

$$\beta_{u,v,p} = \left(R_{u,v}t - t \sum_{j=1}^{N} \mathbf{A}_R[u, v, j] - \sum_{j=1}^{N} \mathbf{M}_B[u, v, j] - 3084 \right)$$

$$\mathbf{T}^{MHM}[u, v, p] = \frac{\sum_{j=1}^{p} \mathbf{M}_{B}[u, v, j] + 3084}{R_{u,v} - \sum_{j=1}^{p-1} \mathbf{A}_{R}[u, v, j]}$$



$$\mathbf{M}_{B}[u, v, p] + \mathbf{A}_{R}[u, v, p] \frac{\sum_{j=1}^{p-1} \mathbf{M}_{B}[u, v, j] + 6001}{R_{u,v} - \sum_{j=1}^{p-1} \mathbf{A}_{R}[u, v, j]} = \mathbf{A}_{B}[u, v, p]$$

$$\beta_{u,v,p} = \left(R_{u,v}t - t\sum_{j=1}^{p-1} \mathbf{A}_R[u,v,j] - \sum_{j=1}^{p-1} \mathbf{M}_B[u,v,j] - 3084 \right)^+$$

$$\mathbf{T}^{MHM}[u, v, p] = \frac{\sum_{j=1}^{p} \mathbf{M}_{B}[u, v, j] + 3084}{R_{u,v} - \sum_{j=1}^{p-1} \mathbf{A}_{R}[u, v, j]}$$



58



ie 59

ТЛП



In such a situation, the MHM leads to a waste of resources The buffer budget will never be used!

because the rate blocks acceptance of other flows

A solution is to artificially reduce the buffer budget!





time

This also reduces the delay of the queue, And hence the lower priority queues can have

- a lower delay, or
- a higher burst budget, or

data

- a higher data rate budget

delay/burst/data rate budget trade-off

ТШ



The resource allocation algorithm is responsible for **adjusting a priori**, the trade-off between resources

The quality of this choice depends on the type of flows

 \rightarrow bursty traffic? rate demanding traffic? low delay?

delay/burst/data rate budget trade-off

Can we do this differently?



We have to find an upper bound independent of the network state...

$$\mathbf{T}[u, v, p] = \frac{\sum_{j=1}^{p} \mathbf{U}_{B}[u, v, j] + \max_{p+1 \le j \le Q_{u,v}} \{l_{u,v,j}^{max}\} + l_{u,v,p}^{max}}{R_{u,v} - \sum_{j=1}^{p-1} \mathbf{U}_{R}[u, v, j]}$$

The MHM does this by bounding $\mathbf{A}_R[u,v,p]$, $\mathbf{M}_B[u,v,p]$ and $l^{max}_{u,v,p}$

The resource allocation algorithm can rather bound the delay itself

$$\mathbf{T}[u, v, p] = \frac{\sum_{j=1}^{p} \mathbf{U}_{B}[u, v, j] + \max_{p+1 \le j \le Q_{u,v}} \{l_{u,v,j}^{max}\} + l_{u,v,p}^{max}}{R_{u,v} - \sum_{j=1}^{p-1} \mathbf{U}_{R}[u, v, j]} \le \mathbf{T}^{TBM}[u, v, p]$$

and let everything vary as long as

 $\mathbf{T}[u, v, p] \le \mathbf{T}^{TBM}[u, v, p] \qquad \forall (u, v, p)$

 $B_{max}(u,v,p) \le \mathbf{A}_B[u,v,p] \qquad \forall \ (u,v,p)$

The resource allocation algorithm can rather bound the delay itself



$$\mathbf{T}[u, v, p] = \frac{\sum_{j=1}^{p} \mathbf{U}_{B}[u, v, j] + \max_{p+1 \le j \le Q_{u,v}} \{l_{u,v,j}^{max}\} + l_{u,v,p}^{max}}{R_{u,v} - \sum_{j=1}^{p-1} \mathbf{U}_{R}[u, v, j]} \le \mathbf{T}^{TBM}[u, v, p]$$
and lot overything vary as long as

and let everything vary as long as

allocation algorithm

$$\mathbf{T}[u, v, p] \le \mathbf{T}^{TBM}[u, v, p] \qquad \forall (u, v, p) \qquad B_{max}(u, v, p) \le \mathbf{A}_B[u, v, p] \qquad \forall (u, v, p)$$

$$B_{max}(u,v,p) = \mathbf{U}_B[u,v,p] + \mathbf{U}_R[u,v,p] \frac{\sum_{j=1}^{p-1} \mathbf{U}_B[u,v,j] + \max_{p+1 \le j \le Q_{u,v}} \{l_{u,v,j}^{max}\} + l_{u,v,p}^{max}}{R_{u,v} - \sum_{j=1}^{p-1} \mathbf{U}_R[u,v,j]}$$

$$\leq \mathbf{A}_B[u,v,p]$$
 \frown buffer capacity



Requires to check lower priority queues and higher priority queues before the addition of a new flow

The resource allocation algorithm can rather bound the delay itself



$$\mathbf{T}[u, v, p] = \underbrace{\sum_{j=1}^{p} \mathbf{U}_{B}[u, v, j] + \underbrace{1542}_{R_{u,v,p}} + l_{u,v,p}^{max}}_{R_{u,v} - \underbrace{\sum_{j=1}^{p-1} \mathbf{U}_{R}[u, v, j]} \leq \mathbf{T}^{TBM}[u, v, p]$$
and let everything vary as long as
$$\mathbf{T}[u, v, p] \leq \mathbf{T}^{TBM}[u, v, p] \quad \forall (u, v, p)$$

$$B_{max}(u, v, p) \leq \mathbf{A}_{B}[u, v, p] \quad \forall (u, v, p)$$

$$B_{max}(u, v, p) = \mathbf{U}_{B}[u, v, p] + \mathbf{U}_{R}[u, v, p] \underbrace{\sum_{j=1}^{p-1} \mathbf{U}_{B}[u, v, j] + \underbrace{1542}_{R_{u,v,p}} + l_{u,v,p}^{max}}_{R_{u,v} - \underbrace{\sum_{j=1}^{p-1} \mathbf{U}_{R}[u, v, j]} \leq \mathbf{A}_{B}[u, v, p] \quad \mathbf{buffer capacity}$$

Requires to check lower priority queues and higher priority queues before the addition of a new flow

because there might be **unknown** besteffort traffic in the lowest priority queue

The Threshold-based Model (TBM)



The Threshold-based Model (TBM)



But no a priori choice on the burst/rate/delay trade-off









data

 $\mathbf{A}_B[u,v,1]$

 $\mathbf{U}_B[u,v,1]$



 $\mathbf{A}_T[u, v, 3]$

time

low priority queue






























Evaluation of the models

Evaluation of the MHM in a **real wind park setup**



Model running on top of OpenDaylight



Evolution of the **packet delay** for the SWO-SW2-SWO flow



Delay border never violated, no packet loss

Delay guarantee 12ms



Evolution of the **packet delay** for the SWO-SW2-SWO flow



Delay border never violated, no packet loss

Delay guarantee 12ms



Simulation of the MHM and TBM

ПП

4 queues, various topologies, various routing procedures, various delay constraints



TBM around $Q_{u,v}$ times slower

TBM potential to perform better but depends on how the routing and resource allocation algorithms avoid the blocking problem

The performance of the models is highly dependent on the other routing/resource allocation algorithms





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Unicast QoS Routing Algorithms for SDN: A Comprehensive Survey and Performance Evaluation

Jochen W. Guck, Amaury Van Bemten, Martin Reisslein, Fellow, IEEE, and Wolfgang Kellerer, Senior Member, IEEE

Abstract-A variety of communication networks, such as industrial communication systems, have to provide strict delay guarantees to the carried flows. Fast and close to optimal quality of service (QoS) routing algorithms, e.g., delay-constrained leastcost (DCLC) routing algorithms, are required for routing flows in such networks with strict delay requirements. The emerging software-defined networking (SDN) paradigm centralizes the network control in SDN controllers that can centrally execute QoS routing algorithms. A wide range of QoS routing algorithms have been proposed in the literature and examined in individual studies. However, a comprehensive evaluation framework and quantitative comparison of QoS routing algorithms that can serve as a basis for selecting and further advancing QoS routing in SDN networks is missing in the literature. This makes it difficult to select the most appropriate QoS routing algorithm for a particuselect the most appropriate Qos routing algorithm for a particu-lar use case, e.g., for SDN controlled industrial communications. We close this gap in the literature by conducting a comprehen-sive up-to-date survey of centralized QoS routing algorithms. We introduce a novel four-dimensional (4D) evaluation framework for QoS routing algorithms, whereby the 4D correspond to the type of topology, two forms of scalability of a topology, and the tightness of the delay constraint. We implemented 26 selected DCLC algorithms and compared their runtime and cost inefficiency within the 4D evaluation framework. While the main conclusion of this evaluation is that the best algorithm depends on the specific sub-space of the 4D space that is targeted, we identify two algorithms, namely Lagrange relaxation-based aggregated cost (LARAC) and search space reduction delay-cost-constrained routing (SSR+DCCR), that perform very well in most of the 4D evaluation space

Index Terms—Delay-constrained least-cost (DCLC) routing, performance evaluation framework, quality of service (QoS), scalability, software-defined networking (SDN).

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I. INTRODUCTION

A. Topic Area: Routing Algorithms for QoS Networking CUTING, i.e., determining a route (path) from a source node to a destination node through a sequence of intermediate switching nodes, is an elementary function of the network layer in communication networks, a diverse array of routing algorithms have been designed. Many routing algorithms have been specifically designed for specific networks settings or applications, see Section I-C.

Providing quality of service (QoS) is an important requirement for a wide range of communication network settings and applications. For instance, multimedia network applications require QoS from the network service, as do many network applications in industrial networks [11] and the smart grid [2] as well as networked control systems [3]. The required QoS is often in the form of delay bounds (constraints) for the data packets traversing the network. Accordingly, extensive research has developed routing algorithms that satisfy given delay constraints while minimizing some cost metric, i.e., socalled delay-constrained least-cost (DCLC) routing algorithms that support QoS networking are often referred to as *QoS routing* algorithms.

Generally, the route determination (computation) is either carried out in distributed nodes, e.g., the control modules in individual distributed Internet Protocol (IP) routers, or by a centralized controller, e.g., a Software-Defined Networking (SDN) controller (4)–8). Distributed routing algorithms had been intensely researched for traditional IP routing, e.g., [9–11], and more recently for a hoc networks, see [12]–[16]. In the mid 1990s, the development of QoS paradigms for the Internet, see [17]–[22], led to a renewed interest in examining routing and spured the development of a plethon of QoS routing algorithms, which mainly targeted distributed computation. In Sharp contrast, the emergence of the Software-Defined Networking (SDN) paradigm [23], [24] has shifted the research focus to centralized network control, including centralized routing computations [25]–160]. The purIEEE TRANSACTIONS ON NETWORK AND SERVICE MANAGEMENT, VOL. 14, NO. 4, DECEMBER 2017

1003

DetServ: Network Models for Real-Time QoS Provisioning in SDN-Based Industrial Environments

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Abstract-Industrial networks require real-time guarantees for the flows they carry. That is, flows have hard end-to-end delay requirements that have to be deterministically guaranteed. While proprietary extensions of Ethernet have provided solutions, these often require expensive forwarding devices. The rise of software-defined networking (SDN) opens the door to the design of centralized traffic engineering frameworks for providing such real-time guarantees. As part of such a framework, a network model is needed for the computation of worst-case delays and for access control. In this paper, we propose two network models based on network calculus theory for providing deter-ministic services (DetServ). While our first model, the *multi-hop* model (MHM), assigns a rate and a buffer budget to each queue in the network, our second model, the threshold-based model (TBM), simply fixes a maximum delay for each queue. Via a packet-level simulation, we confirm that the delay bounds guaranteed by both models are never exceeded and that no packet loss occurs. We further show that the TBM provides more flexibility with respect to the characteristics of the flows to be embedded and that it has the potential of accepting more flows in a given network. Finally, we show that the runtime cost for this increase in flexibility stays reasonable for online request processing in industrial scenarios.

Index Terms—Access control, real-time, industrial network, network modeling, network calculus, quality of service (QoS), software-defined networking (SDN).

I. INTRODUCTION

A. Motivation: Industrial Networking Quality of Service

TNUSTRIAL communications (e.g., machine-to-machine (M2M) communications or production facilities networks) have strict Quality of Service (QoS) requirements, mainly in istic model in the centralized control plane allows to avoi terms of end-to-end delay [1]. This means that flows have endto-end delay bounds that must not be exceeded. In this article, such flows are referred to as *real-time flows*. A wide range of proprietary solutions of Ethernet [3] have been developed for providing this strict QoS. However, these *C. Contribution: DetServ: Network Models for*

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solutions typically require changes within the network protocol stack or impose restrictions on the topology that can be deployed, which leads to expensive forwarding devices.

B. Basis: Centralized Frameworks Based on Software-Defined Networking

Software-Defined Networking (SDN) is a new networking paradigm that runs control functions on a centralized controller which is then able to program the Ethernet forwarding elements in the network using a standardized interface such as OpenFlow [4]. This central view offered by SDN allows to perform traffic engineering based on the global knowledge of the network. Because it only requires simple commodity SDN forwarding elements that can be changed and updated independently [5], SDN is considered as an incepensive solution. Therefore, as elaborated in Section II, a plethora of work has been considering the usage of SDN for the provisioning of QoS [6]–[18]. However, the QoS control provided by these approaches is either too *inaccurate or slow* for industrial applications [18].

As initiated by Jasperneite et al. [19], Guck et al. [10–[18] propose to overcome the two above-mentioned shortcomings by using network calculus, a mathematical modeling framework (introduced in Section III), to maintain a deterministic model of the network state in the control plane. First, network calculus being a deterministic framework, accurate bounds can be computed on a per-flow basis. Second, keeping a deterministic model in the centralized control plane allows to avoid the QoS control loop to go through the forwarding plane, thereby allowing to quickly provision new flow requests [17]. As such, the two drazbacks of existing amonaches are overcome.

C. Contribution: DetServ: Network Models for Deterministic Worst-Case Delay Computation and Access Control

As elaborated in Section IV, a centralized industrial QoS framework requires a network model for the computation of worst-case delays and for access control. The core contribution of this article consists of two network models that can be used as part of such QoS frameworks for providing determin-

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ТЛП

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