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Flow Control For Gigabit Ethernet

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- The presentation discusses various issues related to flow control for Gigabit Ethernet
- XOFF(time) and the PHY based signaling suggested in the buffered repeater proposal are reviewed
- Two advanced (frame based) flow control schemes are then presented:
 - An absolute credit based scheme (as presented in January, modified for Gigabit Ethernet)
 - Simple (we hope -) rate based scheme
- We conclude with some suggestions regarding future work that is needed
 - ... as well as some future/present work which is not needed

XOFF(t) Scheme

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- Concept:
 - An XOFF(t) keeps the link partner quiet for t*slot_time
 - An XOFF(t2) received after an XOFF(t1) overwrites the previous value
 - Thus an XOFF(0) means XON
- This scheme IS the choice made by 802.3x for flow control in full duplex links
 - There was an alternative PHY based scheme
 - It was decided to adopt the frame based scheme
 - ... for many good reasons (should we repeat them here ?)
 - The buffered repeater proposal stated that things are different for Gigabit Ethernet

Well ... are they ?

Latency Model (Frame Based)



Latency Elements (Frame Based)

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 $L_{FC} = Initiate_{Dly} + (2) Link_{Dly} + Msg_{Len} + Stop_{Dly}$

- Initiate_{Dly} = Delay before FC message can be transmitted
- Link_{Dly} = Propagation Delay (1st bit) from MII to MII
- Msg_{Len} = Length of FC message (64 bytes + Preamble)
- Stop_{Dly} = Max number of bits transmitted after receiving last bit of FC frame

Latency Results (Frame Based)

	<u>Bytes</u>
Initiate _{Dly} = 1 max length packet + IPG =	1530
Link _{Dly} = XMT + Link (2Km) + RCV =	
3 + 1250 + 3 =	1256
Msg _{Len} = Min length packet =	72
Stop _{Dly} = 512 b - IPG + 1 max length packet =	1570
Link _{Dly} =	<u>1256</u>
Latency _{FC} =	5684

Latency Model (PHY Based)



Latency Elements (PHY Based)

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 $L_{FC} = Initiate_{Dly} + (2) Link_{Dly} + Sync_{Dly} + Stop_{Dly}$

- Initiate_{Dly} = Delay before Busy signal can be transmitted
- Link_{Dly} = Propagation Delay from MII to MII
- Sync_{Dly} = MAC detecting CRS over the MII
- Stop_{Dly} = Max number of bits transmitted after receiving Busy signal

Latency Results (PHY Based)

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	<u>Bytes</u>
Initiate _{Dly} = 1 max length packet =	1518
Link _{Dlv} = XMT + Link (2Km) + RCV =	
3 + 1250 + 3 =	1256
Stop _{Dly} = 1 max length packet =	1518
Sync. _{Dly} = similar to 100Base-T =	2
Link _{Dly} =	1256
Latency _{FC} =	5550
Summary: 5550 bytes delay for PHY Based Scheme 5684 bytes delay for Frame Based Scheme	

Difference: 134 extra bytes of buffering

.... doesn't look too bad -)

Overhead Computations



- Let's take a look at the *(unreasonable)* worst case scenario (not our idea ... but):
 - 8K bytes buffering per port in the switch (oops ... we meant buffered repeater)
 - High watermark low watermark = 1.5K bytes
 - As soon as we reach the low watermark the following pattern repeats *endlessly:*
 - Congestor sends 1518 bytes packet and stops *exactly* then
 - Congestee *(immediately)* sends an XOFF (64bytes + ipg)
 - Congestee (immediately) empties an 1.518 bytes packet to reach low watermark
 - Congestee *(immediately)* sends an XON (64bytes + ipg)
- ===> Overhead will be (76+76)/(1518+1518) = 5%

Overhead Conclusion

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- This (unreasonably) assumes a worst case condition for a long period of time
 - Congestor pattern is not likely to repeat too often
 - Congestee is extremely unlucky
 - All these (immediately)'s are not too likely to ever happen
- It ignores the *t* in the XOFF(*t*) command
- Given a more practical buffer size of 16K bytes per port (not too expensive for Gigabit ...):
 - High watermark low water mark = 9.5K bytes
 - Overhead (yes, in this ... scenario) is less than 0.8%



===> In ALL *practical* scenarios - overhead is *negligible*

The Issue of Robustness

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- Given a bit error rate of 10**(-12)
 - The chances of a corrupted XOFF(t) frame are:
 - 1 $((1-(10^{**}(-12)))^{**}512) = 0.00000000512$
 - One way to look at it one flow control frame out of every 1953125000 flow control frames is corrupted
 - Assuming the worst case scenario of the overhead computation above, an XOFF(*t*) flow control frame is transmitted every 1518+1518+76 +76 bytes or every 25504 bits ... or 39210 frames per second
 - ==> we will have an XOFF dropped every 1953125000/39210 seconds = 49812 seconds = 830 minutes = every ~14 hours
 - This means ... 2 to 5 packets are dropped (until the "congestee" sends another XOFF)
 - Many more DATA frames will be lost simply due to ... bit error rate

We "think" we can live with 2-5 packets being dropped every 14 hours in this unreasonable worst case scenario -)

Other Observations

- Assuming PHY is full duplex why stop when receiving or for that matter - why look at CRS at all ?
 - ===> The suggested buffered repeater scheme introduces (unnecessary) 50% overhead
 - ... But if this is full duplex, then this is no longer a (buffered) repeater (I guess...)
 - And there are "other" issues:
 - Why have three types of MAC ?
 - Why have PHY and Frame based flow control? One of our (802.3x) goals was to have one flow control for all technologies and all speeds
 - Why specify the architecture and/or behavior of a *switch*?
 - Why does 802.3 need to discuss products ?
 - Why make the PHY more complex (even slightly) ?
 - What does this *"thing"* buy us if "it does not replace traditional CSMA/CD ?"

On Idle/Busy PHY Signals

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- It can be claimed about Idle/Busy:
 - "Let's just define these, maybe someone will use them some day"
 - Problem:
 - As soon as we define them we have to specify their semantics
 - As soon as we specify their semantics we have a new flow control scheme

Do we really need two flow control schemes for Ethernet ?

When the benefit is so small (i.e. less than 134 bytes of buffering)

An Alternative Buffered Repeater Proposal

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Buffered repeater is a *black box* whose architecture is NOT specified by 802.3

A low-cost, high-performance buffered repeater is:

Trivial to build using: 1. 802.3x frame based flow control 2. FDX MAC as defined by 802.3x

No need to define a new type of MAC No need for defining (and giving semantics) to idle/busy

Why spend so much time and energy and add complexity and definitions and semantics for something that *might* buy us 134 bytes of buffering in worst case conditions?

... In short, we do not need an alternative PHY based flow control for Gigabit Ethernet

Buffered Repeater Conclusion

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====> There is full duplex Ethernet and there is CSMA/CD Ethernet:

- 1. Buffered repeater as proposed so far does not fit any of these branches:
 - a. For full duplex we simply do not need it (it adds nothing except additional complexity)
 - b. For half duplex (i.e. CSMA/CD) it is not relevant at all
- 2. Buffered repeater for the full duplex branch can be designed without any aditional work from 802.3z based on 802.3x work
- ===> Instead of taking more of 802.3z's valuable time to discuss switch architecture and Phy based flow control schemes which buy us (at most) 134 bytes of buffering -

We shall invest the time in looking at

- 1. More advanced flow control schemes (sub task force ?)
- 2. Solutions to the CSMA/CD problems (performance, capture effect, etc... another/same sub task force ?)

Credit Based Flow Control Concept

- Absolute credits are sent:
 - A credit(N Bytes) means: "You can send me N bytes"
- Transmit side is trivial:
 - After transmission is started it shall never be stopped
 - A new transmission can start if
 - I have credits to send at least 1500 bytes
 - I have credits to send N bytes & I have a frame with a (known) length <= N
 - When a frame carrying credits is received:
 - Simply write the new value in the credit counter
 - No need to synchronize
 - Value is *guaranteed* to be *accurate* when it is received

Credit Based Flow Control Concept (cont.)

- Receive side is very simple:
 - Assuming there are N free bytes
 - When there is a *need* (see below) or when there is *nothing better to do* (see below):
 - send a credit frame with (N Const) bytes
 - Const = Sending time + RTD + Processing time (more details to follow)



- There is a *need* to send credit when:
 - N constant > some threshold AND
 - Link partner is short with credits OR we didn't send a credit frame for the last TBD time (order of seconds)
- There is nothing better to do means:
 - No transmit frames OR no available credits

Credit Based Flow Control -Constant Part Calculations

- Partner will get a new credit value sent now (time is t) in time t + DT where DT is the sum of:
 - S: The time it took the last byte of the frame to cross MDI (sending end)
 - HRTD: The time it took the last byte of the frame to cross the MAC Control Interface (receiving end)
 - **R**: Processing time (receiving end)
- - **S** is ~ 64 bytes time, **R** is negligible and **HRTD** is either
 - Bounded in 2KM Gigabit links (1250 bytes ?)
 - Can be computed in future extensions
- •
- Bytes in flight: bytes that were sent before time *t* but did not reach receiving end (In Gigabit links < 1250 bytes ?)
- ==> Value to send == # of free bytes C (constant)
 - Where **C** == (**DT** * line rate in bytes) + bytes in flight

Credit Based Flow Control -Intermediate Summary

- Basically this is it:
 - No frame is sent before receiver is ready for it
 - The no frame lost is guaranteed
 - No need to act in real time to prevent frame loss
 - No need for ANY re-synchronization due to frame lost
 - New credit frames turn old/lost frames obsolete
 - Sending a credit after a long (seconds) gap from the last credit sent is all that is needed to deal with lost credits
- It is important to note that the standard needs only to specify the syntax of the credit frame:
 - One new opcode is needed
 - Only the behavior of the transmitting end needs to be specified

Credit Based Flow Control -Pseudo Code

- The next part of the of presentation will provide the pseudo code
 - We believe it might help:
 - To understand this scheme better
 - To show how simple it is
 - To catch bugs and uncovered details
- Most of the code:
 - May be implemented by the MAC client
 - May be implemented by the MAC control sub-layer
- In order to facilitate this approach all we *have* to do is:
 - Specify the needed opcode
 - State that the transmitter is not allowed to send frames if it doesn't have credit to send them

Credit Based Flow Control -Constants

C: integer;	{the constant, whose computation was presented before, to be subtracted from free_bytes when sending credits}	
CRD_MAX_GAP: integer;	{max time which is allowed between credit frames}	
SND_CRD_MIN: integer;	{minimal number of credits to send}	
PRIO_THRSLD: integer;	{when LP has less credit than this number, we set high the priority of returning credits}	
Note: ALL these constants might be implemented by the MAC client		

Credit Based Flow Control -Variables

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byte_count: integer;	{number of bytes we can send}
LP_byte_count: integer;	<pre>{how many bytes link partner can still send us}</pre>
fctl_priority_flag: Boolean	; {when set we need to send credit as soon as possible}
free_bytes: integer;	{how many free bytes do we have being updated by the MAC client}
crd_time_sent: integer;	{the last time we sent a credit frame}

Note: ALL these variables except for byte_count might be implemented by the MAC client

Credit Based Flow Control -Transmitter Process

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Transmits credit or normal frames according to availability of credits and the priority of sending credits

```
Loop forever
```

{

If (fctl_priority_flag && (free_bytes > (SND_CRD_MIN + C))) then
 transmit_fctl(free_bytes - C);

else if (there is a frame to transmit with length < byte_count)

start a new transmission;

while transmitting decrement byte_count every byte sent;

else if (free_bytes > (SND_CRD_MIN + C)) than transmit_fctl(free_bytes - C);

Note: only the part within the green box need to be implemented by the MAC Control sub-layer

Credit Based Flow Control -FCTL Processes & Procedures

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Procedure Transmit_fctl(value);

send a flow control frame with number of credits equal value;

```
LP_byte_count := value;
```

```
crd_time_sent := NOW; fctl_priority_flag := FALSE;
```

Process Fctl_prio

Loop forever

```
every byte received:
```

```
LP_byte_count := LP_byte_count - 1;

If ((LP_byte_count < PRIO_THRSLD) ||

(NOW - crd_time_sent) > CRD_MAX_GAP)

than fctl_priority_flag := TRUE;
```

Note: only the part within the green box need to be implemented by the MAC Control sub-layer

Credit Based Flow Control -Summary

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- A simple (we hope) credit based scheme was presented
- The main features it provides:
 - No frame loss
 - No need for real time response
 - Suits high end and low end implementations (with respect to buffering)
 - Applicable to switch to switch and switch to end node links
- The main feature to emphasize:

Can work even with as little buffering as 4K bytes for 2KM Gigabit connections

Simple Rate Based Scheme -Introduction

- A word of caution:
 - A rate based scheme can turn into a *nightmare* with respect to both implementation and analysis
 - maybe this explains why a rate based scheme was adopted by *"another"* technology -)
 - ==== > the scheme we have in mind is very (too ?) simple
- The main advantages of rate based schemes:
 - Do not produce the "all or none" behavior
 - System behavior is smoother
 - Involve very little computation for ports
 - Are (somewhat?) less affected by the length of the links
- Main disadvantage: frames may be lost

Rate Based Scheme -Concept

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- Basic idea use IPG to control the rate
- Three commands in addition to the XOFF(t):
 - Set_ipg(n): ipg is <u>set</u> to n
 - Slow_down(k): ipg is <u>multiplied</u> by 2**k
 - Speed_up(*I*): ipg is <u>divided</u> by 2**/ (down to current (*minimal*) ipg)

Implementation is ... trivial:

- Instead of using a constant IPG use a programmable value which may be affected by the flow control commands
- An acceptable simplification:
 - Use only the Set_ipg(n) command

Rate Based Scheme -Usage

- Not as trivial as the XOFF(t) or credit based schemes
 - However this is NOT the scope of 802.3
 - ===> All we have to do is provide the hooks and be convinced that they are useful
- It does *NOT* guarantee zero frame loss
- Simplest usage:
 - Define a series of high watermarks H1...Hn
 - Whenever we exceed a watermark we send Slow_down(1)
 - Define a series of low watermark L1 ... Lk
 - Whenever we go under a watermark we send Speed_up(1)
 - Whenever there is a major change we may use
 - XOFF command
 - Set_ipg command to set ipg to a suitable value immediately

Rate Based Scheme -Evaluation

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- Advantages:
 - Does not produce the "all or none" behavior
 - Goes well with the XOFF(t) scheme
 - Simpler to implement than credit based scheme
 - Less (directly) affected by RTD
 - Goes well with flow control of "other technologies"
 - Provides more room for clever system design (i.e., is more flexible) than other schemes
 - Better suits Ethernet "philosophy"

Disadvantages:

- Does not guarantee zero frame loss
- Usage is less straightforward than in credit based or XOFF based schemes
- More difficult to analyze

Summary

