

Leading the Next

# Analysis of Clock Synchronization Approaches for Residential Ethernet

Geoffrey M. Garner  
Kees den Hollander  
SAIT / SAMSUNG Electronics  
[gmgarner@comcast.net](mailto:gmgarner@comcast.net)  
[denhollander.c.j@samsung.com](mailto:denhollander.c.j@samsung.com)

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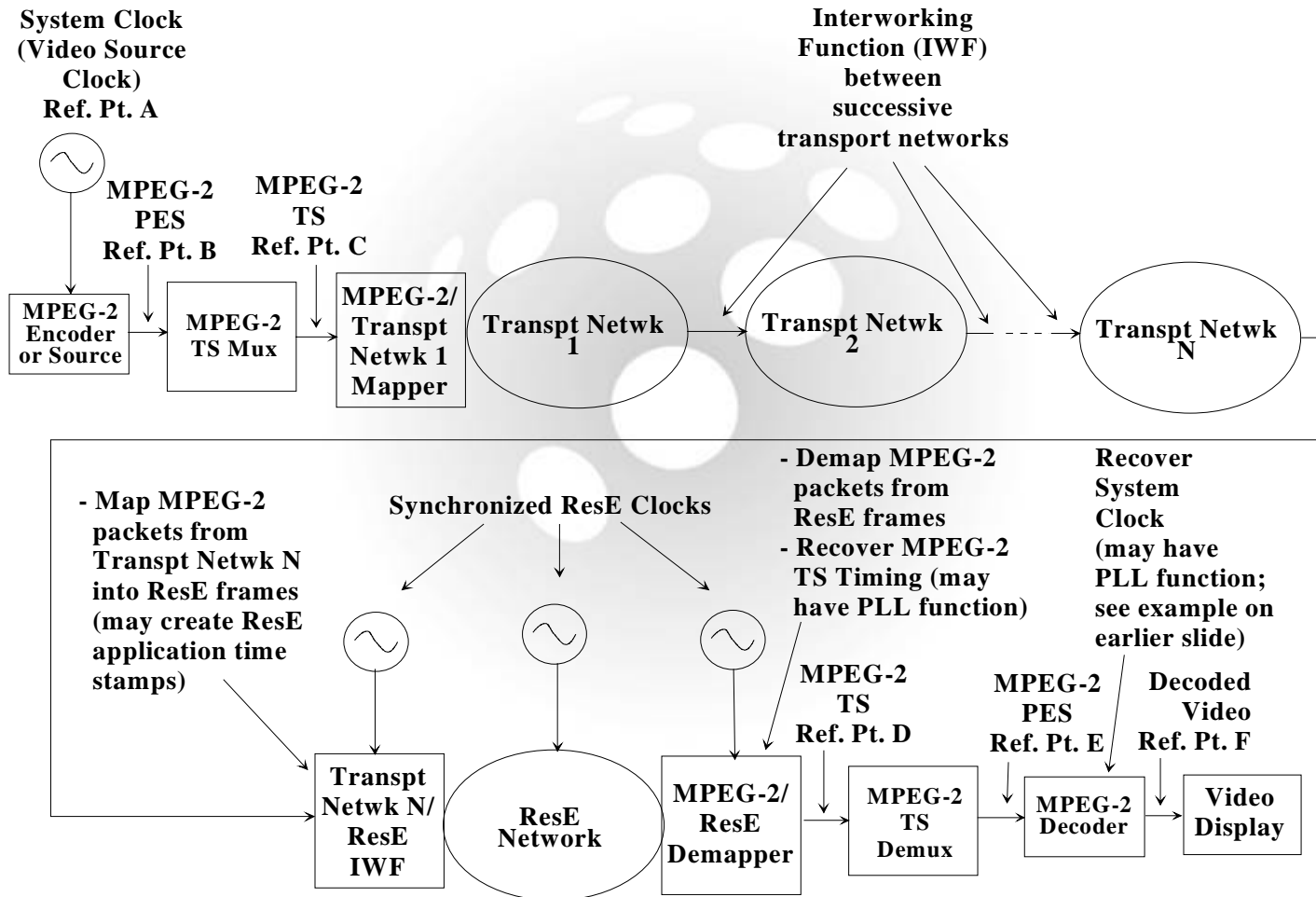
SAMSUNG

- ❑ Introduction
- ❑ Application reference models
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- ❑ Synchronization approaches for ResE
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- ❑ Simulation cases and results
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- ❑ Residential Ethernet (ResE) is a new standardization activity in IEEE 802 that is considering extensions to Ethernet to allow the transport of time-sensitive traffic (e.g., high quality audio and video (A/V))
- ❑ A/V applications have tight jitter and wander requirements that must be met end-to-end
- ❑ To meet these requirements, synchronization is required at ResE ingress and egress points
- ❑ This analysis investigates if and how synchronization approaches based on IEEE 1588 can meet the ResE requirements

# Application Reference Models

## Example Reference Model for Transport of MPEG-2 Video over Service Provider Networks and Residential Ethernet [2]

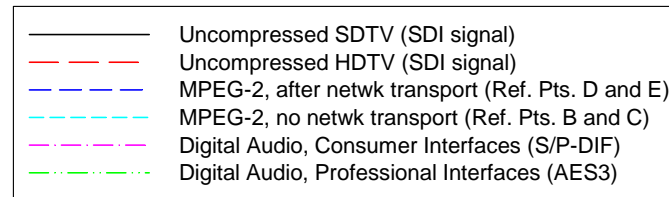


# End-to-End Requirements

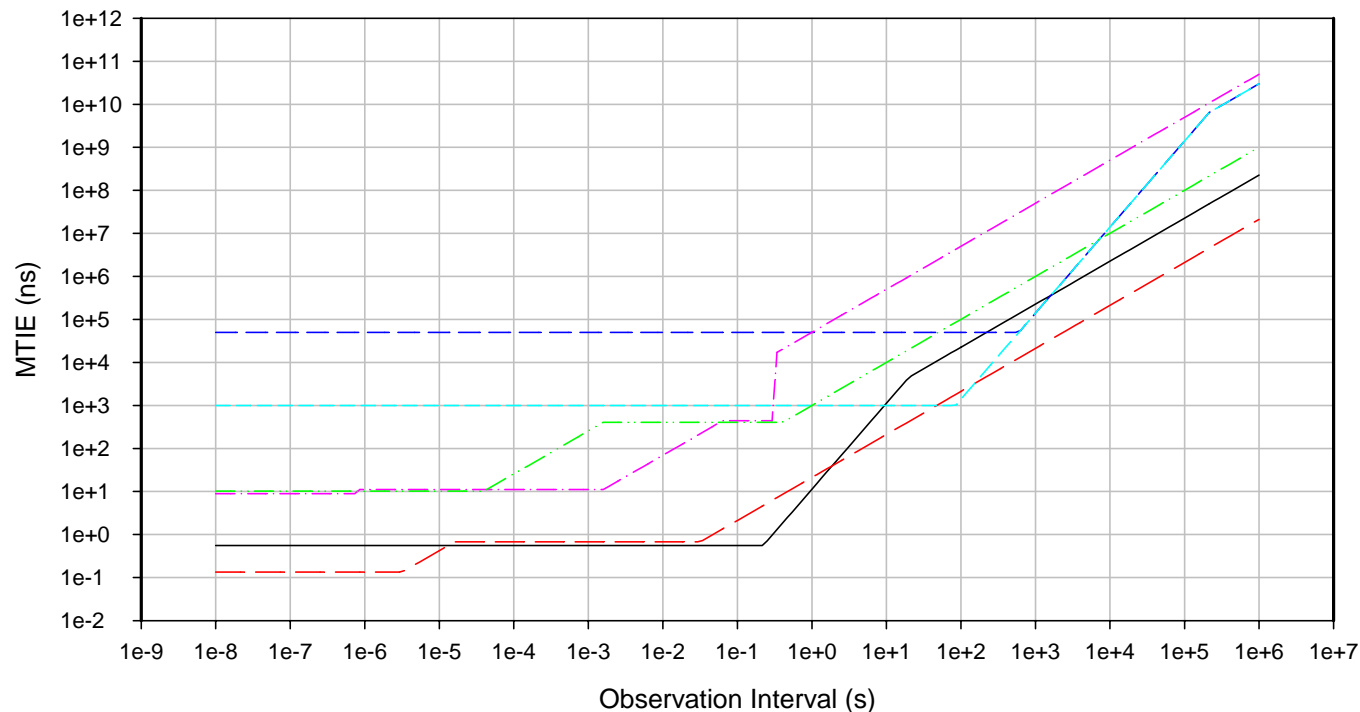
Summary of End-to-End Application Jitter and Wander Requirements  
(see [2] and references given there)

Requirement	Uncompressed SDTV	Uncompressed HDTV	MPEG-2, with network transport	MPEG-2, no network transport	Digital audio, consumer interface	Digital audio, professional interface
Wide-band jitter (UIpp)	0.2	1.0	50 $\mu$ s peak-to-peak phase variation requirement (no measurement filter specified)	1000 ns peak-to-peak phase variation requirement (no measurement filter specified)	0.25	0.25
Wide-band jitter meas filt (Hz)	10	10			200	8000
High-band jitter (UIpp)	0.2	0.2			0.2	No requirement
High-band jitter meas filt (kHz)	1	100			400 (approx)	No requirement
Frequency offset (ppm)	$\pm 2.79365$ (NTSC) $\pm 0.225549$ (PAL)	$\pm 10$	$\pm 30$	$\pm 30$	$\pm 50$ (Level 1) $\pm 1000$ (Level 2)	$\pm 1$ (Grade 1) $\pm 10$ (Grade 2)
Frequency drift rate (ppm/s)	0.027937 (NTSC) 0.0225549 (PAL)	No requirement	0.000278	0.000278	No requirement	No requirement

## End-to-End Application Jitter and Wander Requirements Expressed as MTIE Masks [2] (see Appendix II for MTIE definition)

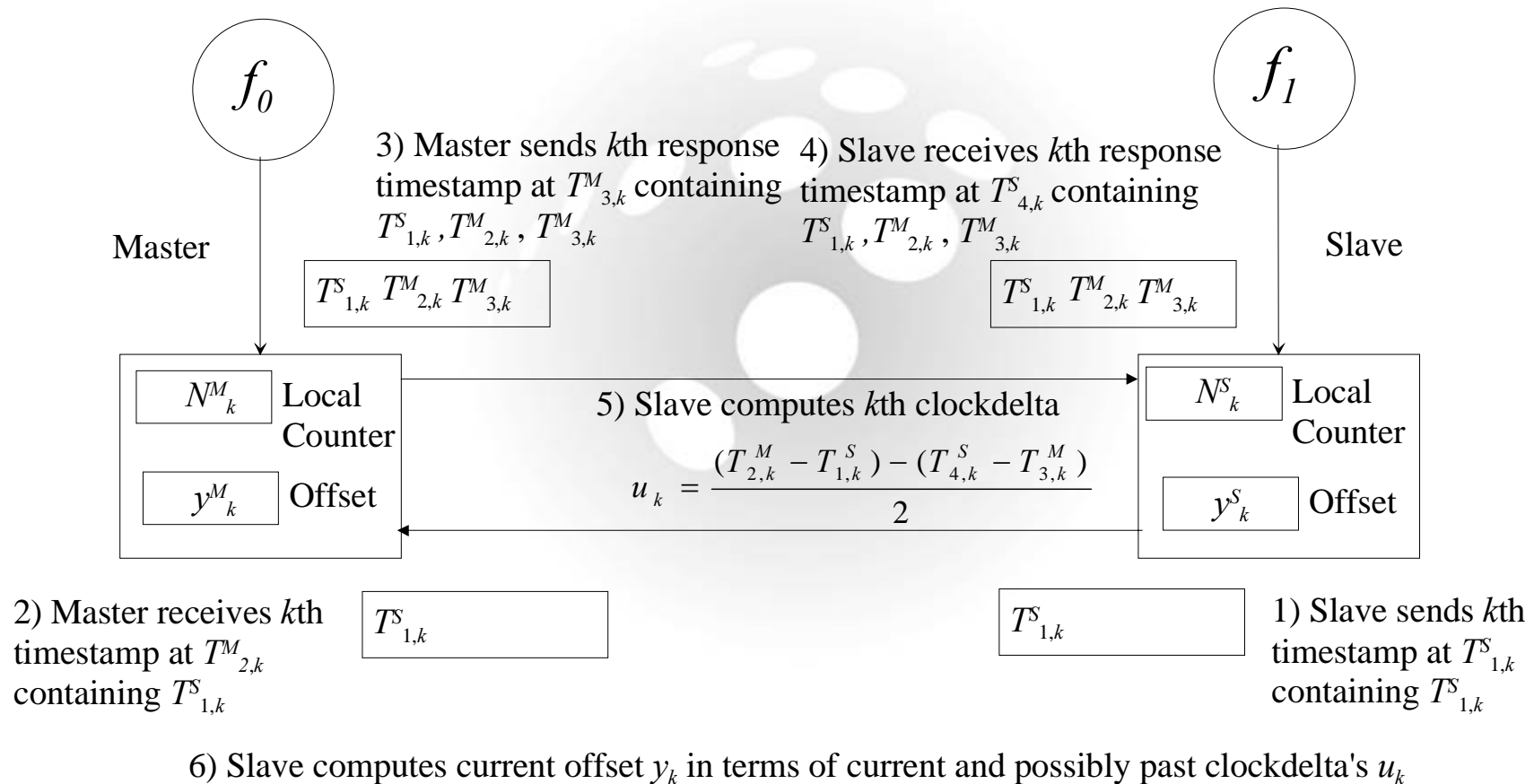


Network Interface MTIE Masks for Digital Video and Audio Signals



## Basic 2-Way Time Stamp Approach used in IEEE 1588

- ResE will use this basic approach; however, a number of variations are possible
- Generally assumed a filtering function will be present at the endpoint
  - May be present at intermediate nodes (i.e., in some variations)

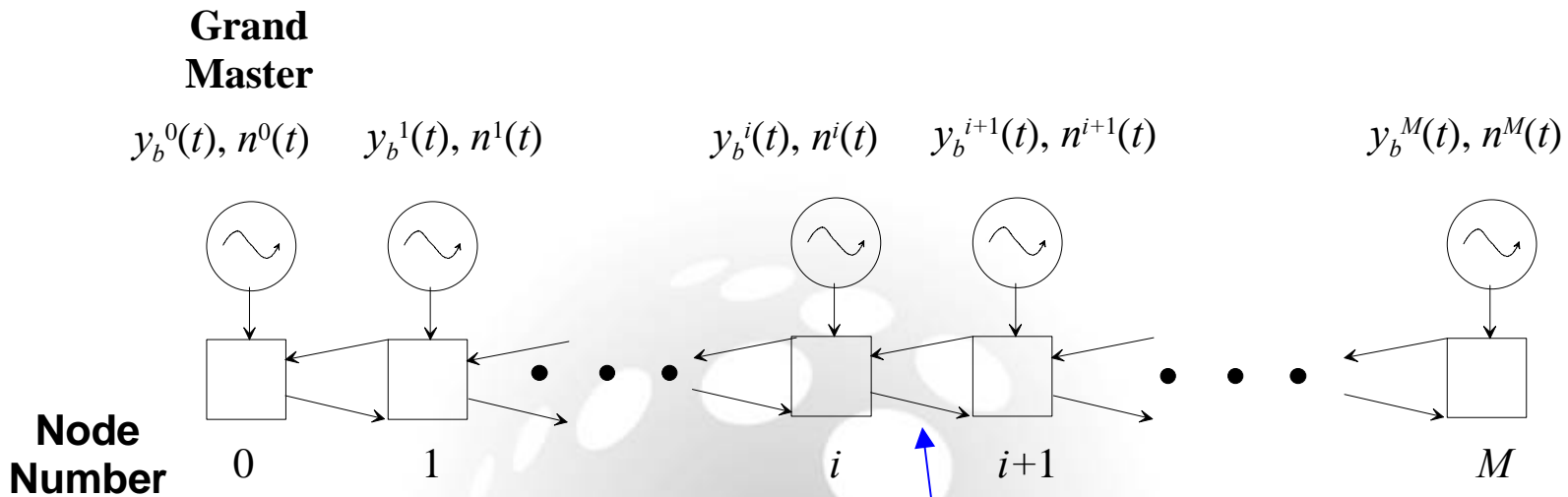


## Variations/Choices

- 1) Use one-way time stamp scheme with less frequent two-way exchange; obtain delay from two-way exchange and assume delay is fixed until next two-way exchange
- 2) **Instantaneous phase adjustments at intermediate nodes**
- 3) **Instantaneous phase and frequency adjustments at intermediate nodes (with instantaneous frequency adjustments possibly less frequent)**
  - Described in [4]
- 4) **Filtered phase adjustments at intermediate nodes, using digital filter running at local clock rate (with or without instantaneous frequency adjustments)**
- 5) Full phase-locked loops (PLLs) at intermediate nodes (i.e., filtered phase and frequency adjustments)
- 6) Use of transparent clock nodes
  - a) End-to-end versus peer-to-peer
  - b) Whether or not to adjust rate of local oscillator in transparent clock and, if so, whether to do filtering
- 7) Time stamp reflects current time versus delay by some number of frames
- 8) Time stamp reflects local free-running clock time versus latest corrected time based on most recent time stamps and possible filtering)



# Synchronization Model



$x_b^i(t)$  = phase offset of clock  $i$  relative to UTC (ns)

$y_b^i(t)$  = frequency offset (pure fraction) of clock  $i$  relative to UTC

$n^i(t)$  = phase noise of clock  $i$  (ns)

$$x_b^i(i) = \int_0^i y_b^i(t) dt + n^i(t) = y_b^i t + n^i(t)$$

- Assumes the frequency offset is constant over time
  - Assumes the phase offsets are zero at  $t = 0$
  - We are interested in timing relative to the GM;
- therefore, can set  $y_b^0(t) = n^0(t) = 0$

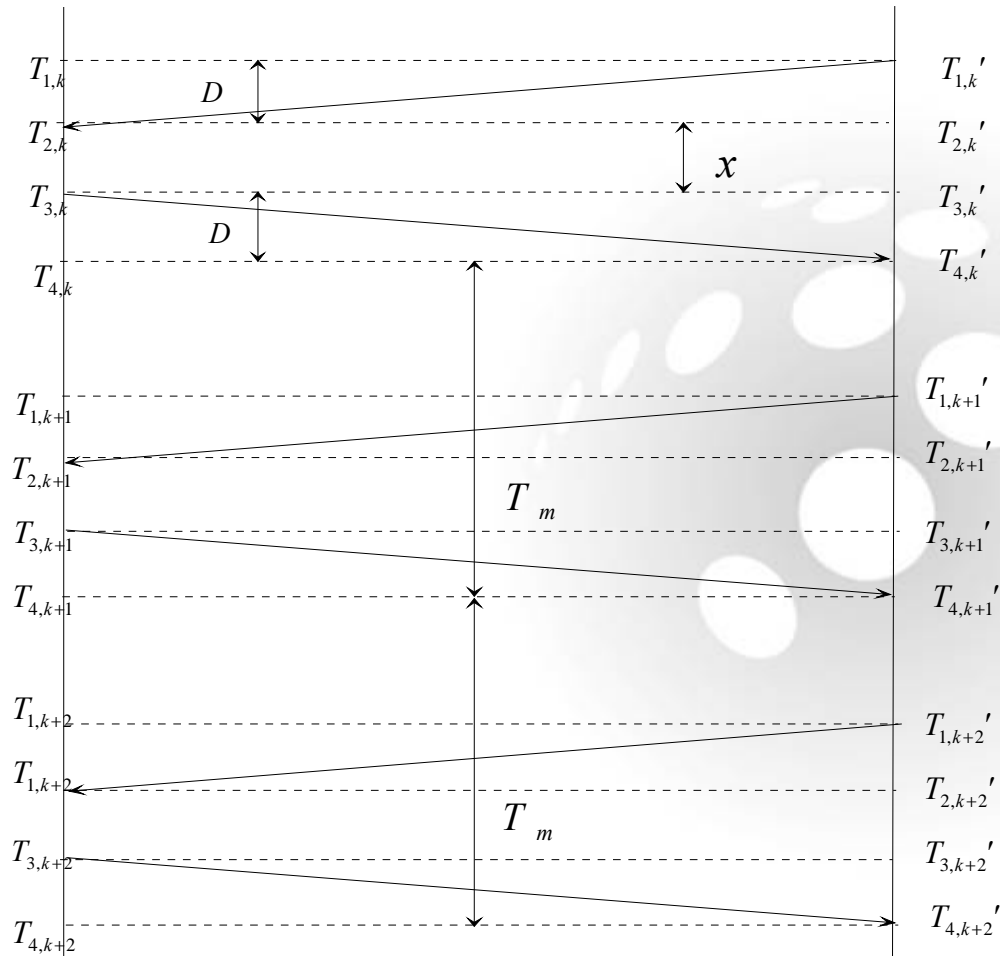
Message Exchanges between clock  $i$  (master) and clock  $i+1$  (slave)

- Note that messages from master to slave and slave to master do not necessarily occur at the same times
- Note that messages from master to slave and slave to master may not occur at the same rates

# Synchronization Model

Clock  $i$   
(master)

Clock  $i+1$   
(slave)



$T_m$  = time between successive messages from master to slave, measured relative to UTC

$D$  = propagation delay between master and slave

$x$  = time offset between master and slave (will be initialized randomly between 0 and  $T_m$  and either kept constant or allowed to change by frequency offset between master and slave multiplied by  $T_m$ )

- Assume  $D \ll T_m$ , and therefore probability that messages from master to slave and slave to master overlap in time is negligible

- Unprimed quantities are relative to master clock

- Primed quantities are relative to slave clock

Then, can express the phase offset in discrete time ( $k$  = time index; UTC time at step  $k = kT_m$ )

$$x_{b,k}^i = y_b^i T_m k + n_k^i$$

## □ Outline of model derivation

- For variations (3) and (4), express frequency offset estimate of slave relative to master over  $P$  time steps in terms of the  $x_{b,k}^i$  and  $x_{b,k}^{i+1}$  (tilde denotes relative frequency offset between current and previous node)
  - Compare time differences in free-running master and slave clocks over  $PT_m$

$$\tilde{y}_{kP}^i = \frac{(PT_m + x_{b,kP}^i - x_{b,(k-1)P}^i) - (PT_m + x_{b,kP}^{i+1} - x_{b,(k-1)P}^{i+1})}{PT_m + x_{b,kP}^{i+1} - x_{b,(k-1)P}^{i+1}}$$

$$\tilde{y}_{kP+1}^i = \tilde{y}_{kP+2}^i = \dots = \tilde{y}_{kP+P-1}^i = \tilde{y}_{kP}^i$$

- For variations (3) and (4), calculate cumulative frequency offset of current node relative to GM  $y_k^i = \sum_{j=1}^i \tilde{y}_k^j$
- For variation (3) and (4), express corrected phase error estimate  $x_j^i$  in terms cumulative frequency offset estimate and free-running clock phase error  $x_{b,j}^i$ 
  - Choose phase error estimate at all time steps between frequency updates to be consistent with current frequency offset estimate

$$\frac{(j - kP)T_m + x_j^i - x_{kP}^i}{(j - kP)T_m + x_{b,j}^i - x_{b,kP}^i} = 1 + y_k^i$$

$$x_j^i = x_{kP}^i + (x_{b,j}^i - x_{b,kP}^i)(1 + y_k^i) + (j - kP)T_m y_k^i$$

- For variations (2) – (4), calculate clock delta in terms of either corrected phase error estimates (for cases where frequency adjustments are made) or free-running clock phase errors
  - Apply result for clock delta in step (5) on slide 7
  - Need phase error values at intermediate times  $T_{1,k}', T_{2,k}, T_{3,k}$
  - Obtain these by interpolation; result depends on  $x$  and  $D$ 
    - Take limit  $D \rightarrow 0$
    - Note: assumption is being made that we can interpolate on the noise
      - Reasonable as long as the desired noise level is chosen for sampling rate  $T_m$
  - See paper for details
- Calculate cumulative clock delta for all nodes up to the current one (GM clock delta is zero)
- Add cumulative clock delta to corrected or free-running clock phase error to obtain unfiltered phase estimate
- Filter the unfiltered phase estimate with a digital filter that runs at the local clock rate

- Since the filter is linear, the result is the same for the case where each clock delta is filtered at each respective intermediate node versus filtering the cumulative clock delta

- If synchronization is needed at each node, the work is the same in either case

- Filter model is a digital implementation of standard 2<sup>nd</sup> order, linear filter with 20 dB/decade roll-off

$$H(s) = \frac{2\zeta\omega_n s + \omega_n^2}{s^2 + 2\zeta\omega_n s + \omega_n^2}$$

$\omega_n$  = undamped natural frequency

$\zeta$  = damping ratio

$$f_n = 3 \text{ dB bandwidth} = (\omega_n / 2\pi) \left[ (2\zeta^2 + 1) + \sqrt{(2\zeta^2 + 1)^2 + 1} \right]$$

$$H_p = \text{gain peaking} = \left[ 1 - 2\alpha - 2\alpha^2 + 2\alpha\sqrt{2\alpha + \alpha^2} \right]^{1/2}$$

where  $\alpha = 1/(4\zeta^2)$

- The digital implementation is obtained by expressing the filter in state variable form (See [6] and [7] for details)

- State vector at current time step is written as convolution integral of input vector and impulse response matrix
  - Impulse response matrix is calculated exactly and integral is evaluated using trapezoidal approximation for input
  - Output is written in terms of states

## □ Additional aspects of model

- Clock noise model is described in appendix
- Simulation time step is a sub-multiple of the inter-message time  $T_m$  (cannot exceed  $T_m$ )
- Time between frequency estimate updates is a multiple of  $T_m$
- Time offset between master→slave and slave→master messages may be initialized randomly or initialized with user-specified values
- Time offset between master→slave and slave→master messages may remain constant over the simulation or vary over  $T_m$  by the relative frequency offset between master and slave, multiplied by  $T_m$ 
  - Former requires that the master and slave send messages at the same rate
  - Latter corresponds to messages being sent at the free-running clock rates
- Finite precision of clock is modeled
  - Granularity, in units of time, is supplied as input parameter

- ❑ 10 hops
  - GM followed by 10 slave clocks, in chain
- ❑ Slave clock frequency tolerance =  $\pm 100$  ppm
- ❑ Filter bandwidth = 10 Hz
- ❑ Filter gain peaking = 0.1 dB
- ❑ Simulation time step = 0.01 ms
  - Used small time step to ensure phase peaks were captured

## □ Assumptions

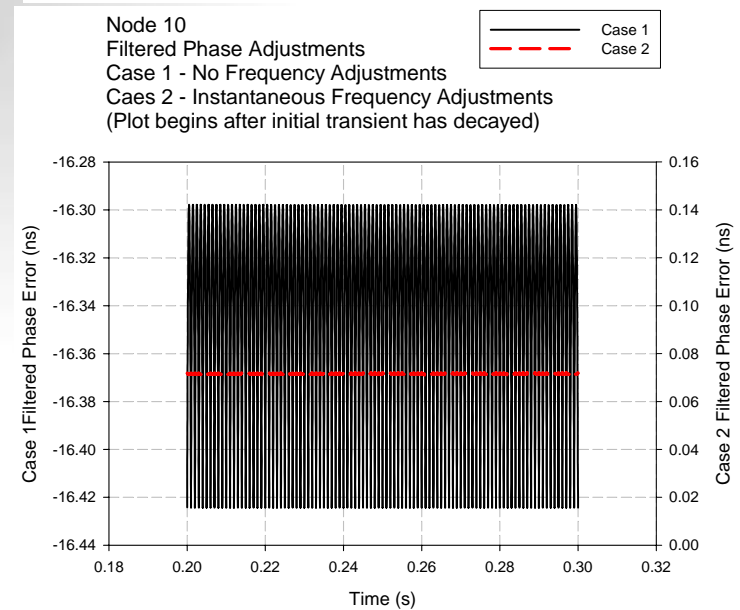
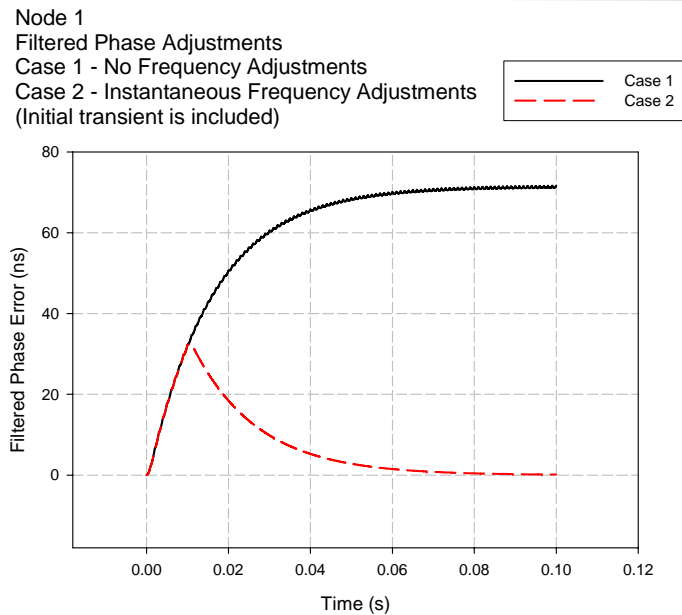
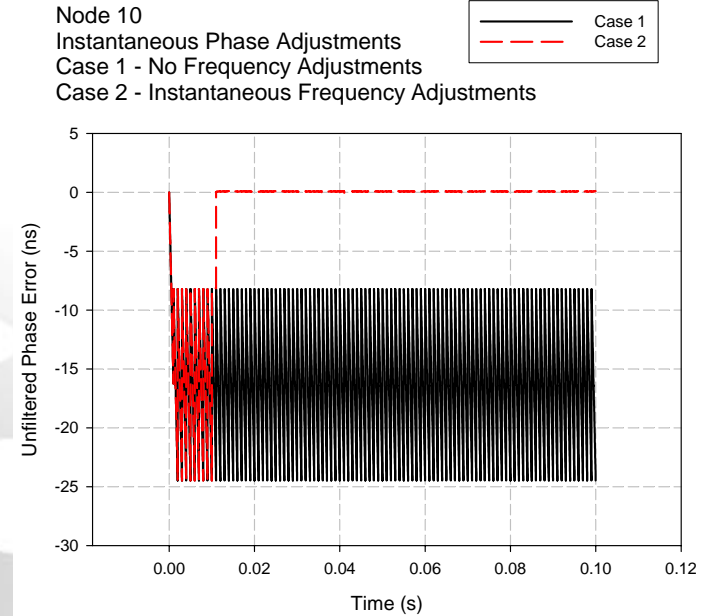
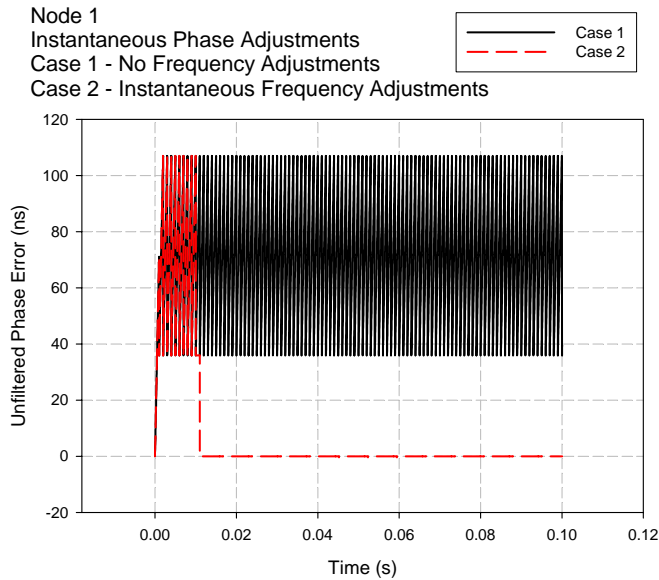
- No clock phase noise
- Granularity of clock = 0
- No frequency adjustments (Case 1); Instantaneous frequency adjustments (Case 2)
  - Inter-message time ( $T_m$ ) = 1 ms
  - Time between frequency offset updates = 10 ms (Case 2)
- Offset between master→slave and slave→master messages set to  $T_m$  at each node (deterministic and constant)

## □ Results (see plots on next slide)

- With instantaneous phase adjustments (no filtering) and no frequency adjustments, steady-state peak-to-peak phase error can be large (tens of ns) and depends on frequency offsets
  - With 10 Hz filter and no frequency adjustments, steady-state peak-to-peak phase error is reduced to a few tenths of a ns
- With instantaneous frequency adjustments, steady state peak-to-peak phase error is very small
  - Approximately 0.07 ns with no filtering
  - Approximately 0.00055 ns (0.55 ps) with filtering
  - With no clock noise and zero phase granularity, frequency offsets can be measured very accurately
- Phase variation does not increase monotonically with number of clocks in chain



# Simulation Cases 1 and 2



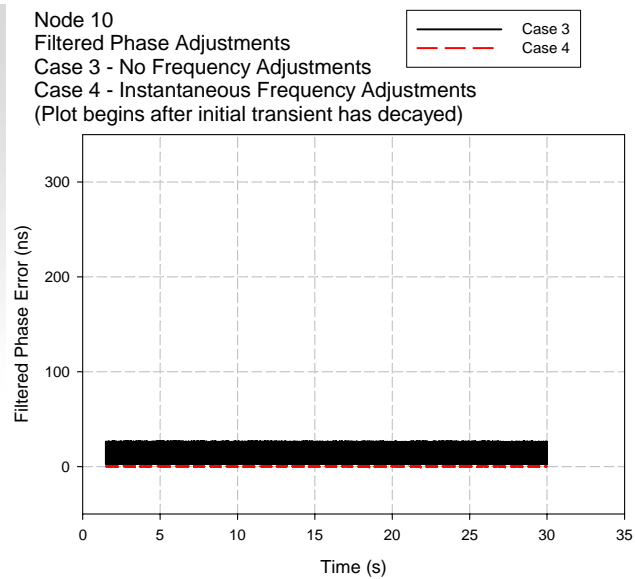
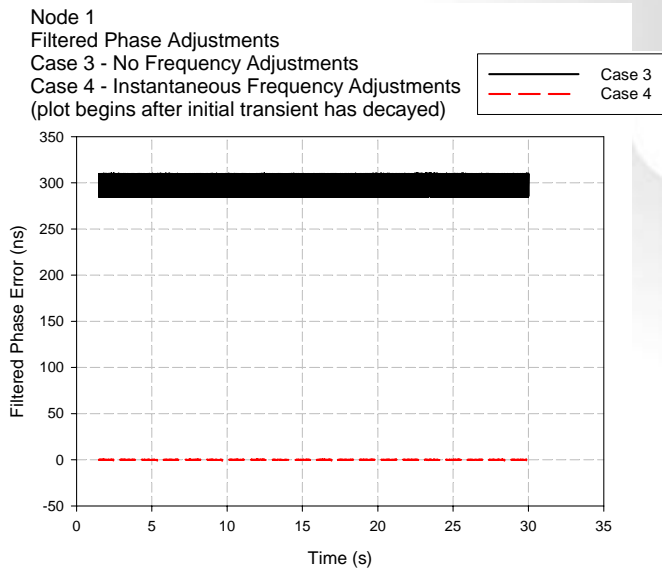
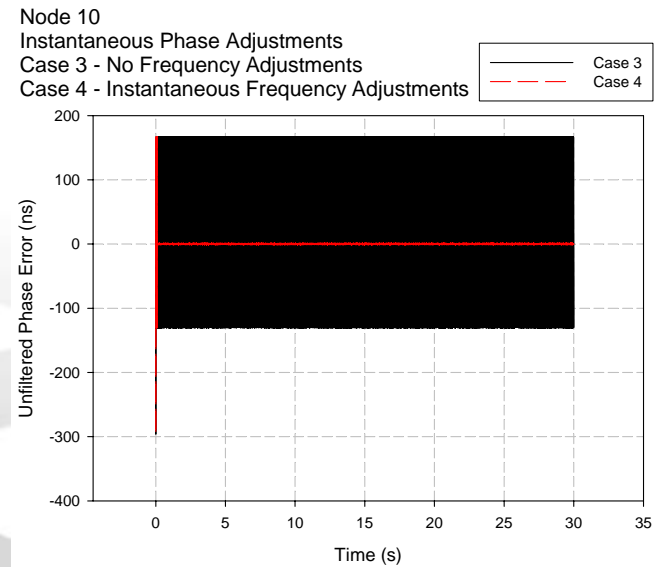
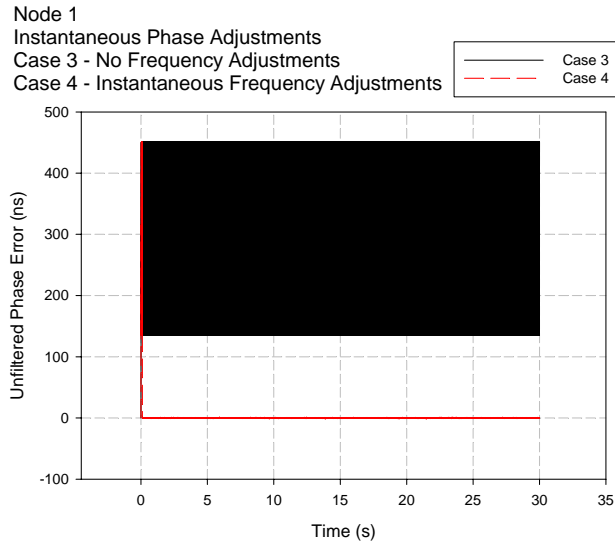
## □ Assumptions

- With clock phase noise (model described in Appendix)
- Granularity of clock = 1 ns
- No frequency adjustments (Case 3); Instantaneous frequency adjustments (Case 4)
  - Inter-message time ( $T_m$ ) = 10 ms (suggested in [4])
  - Time between frequency offset updates = 100 ms (Case 4) (suggested in [4])
- Offset between master→slave and slave→master messages initialized randomly at each node
  - All nodes send messages at the same rate (offsets remain constant over simulation)

## □ Results (see plots on next slide)

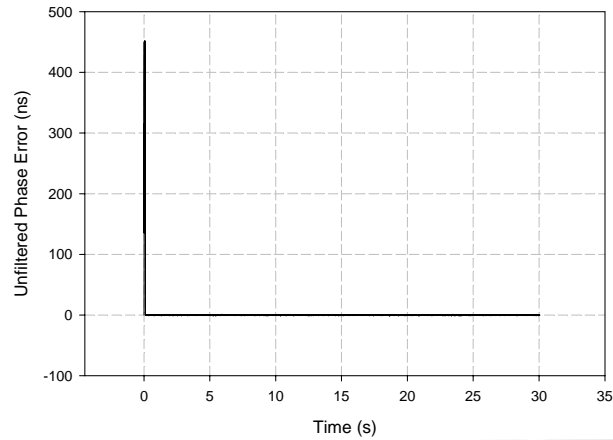
- With 10 Hz filter, MTIE is considerably smaller with frequency adjustments (compared to without frequency adjustments), at longer observation intervals
  - Approximately 1 – 1.5 ns with frequency adjustments
  - Approximately 10 – 50 ns without frequency adjustments
- Without filtering, MTIE ranges from approximately 160 – 600 ns without frequency adjustments and 2 – 4 ns with frequency adjustments
- Phase variation does not increase monotonically with number of clocks in chain (in all cases)
- Note that the results exhibit large statistical variability
  - Must run multiple, independent replications of the simulations to obtain confidence intervals for the results

# Simulation Cases 3 and 4

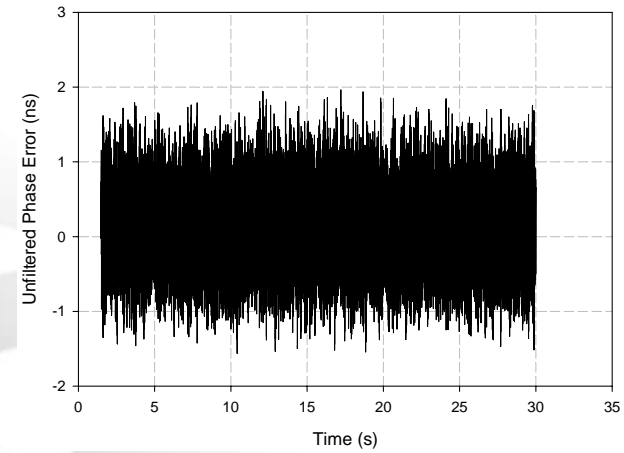


# Simulation Case 4 (Detailed View)

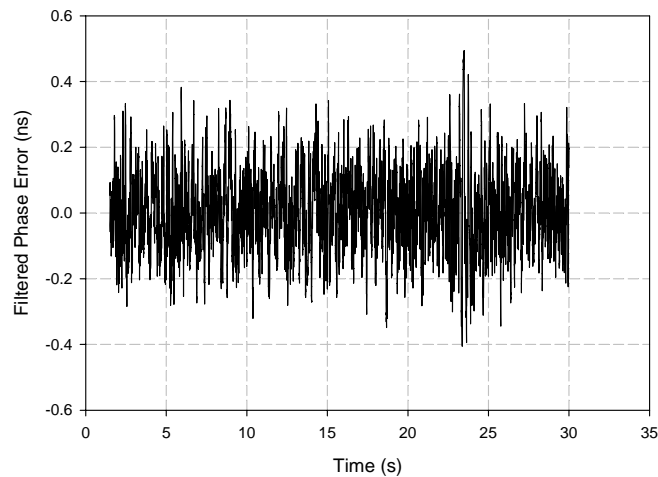
Case 4, Node 1  
Instantaneous Phase Adjustments  
Instantaneous Frequency Adjustments



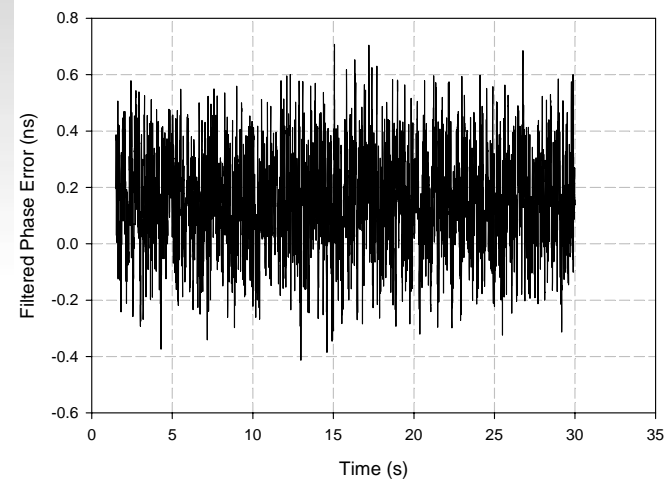
Case 4, Node 10  
Instantaneous Phase Adjustments  
Instantaneous Frequency Adjustments  
(Plot begins after initial transient has decayed)



Case 4, Node 1  
Filtered Phase Adjustments  
Instantaneous Frequency Adjustments  
(plot begins after initial transient has decayed)

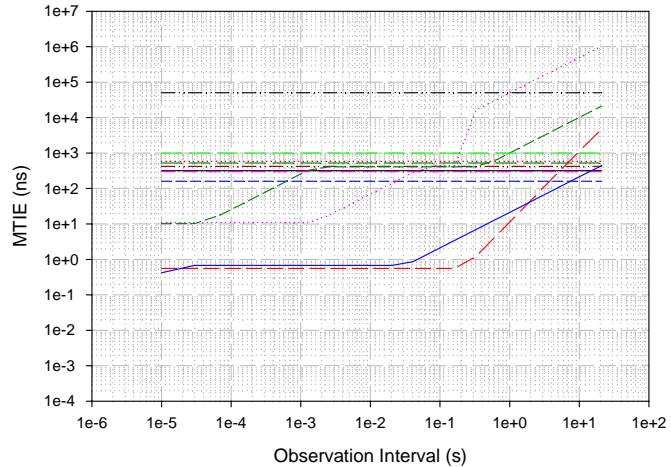


Case 4, Node 10  
Filtered Phase Adjustments  
Instantaneous Frequency Adjustments  
(Plot begins after initial transient has decayed)

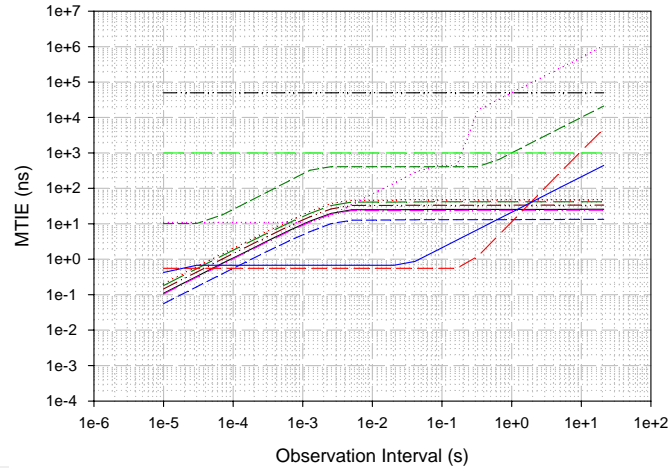


# Simulation Cases 3 and 4

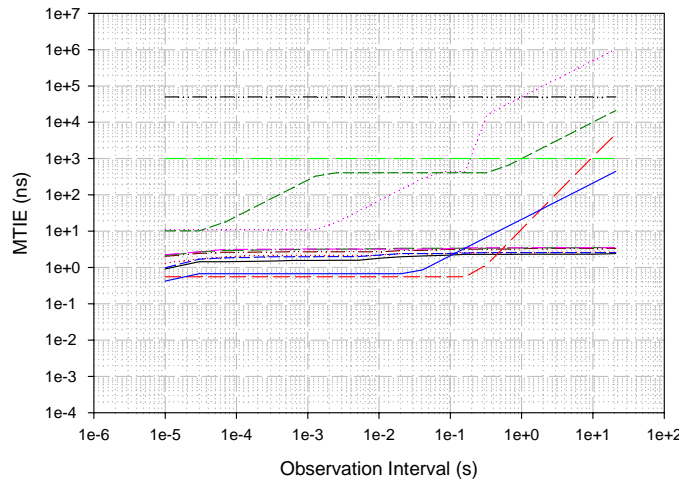
Case 3  
Instantaneous Phase Adjustments  
No Frequency Adjustments



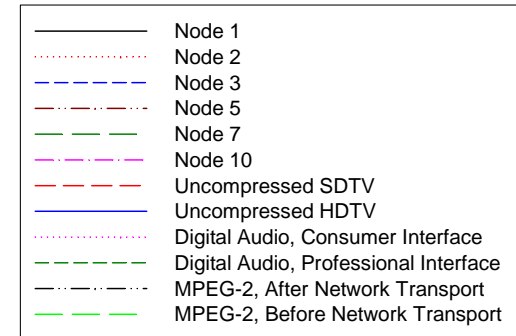
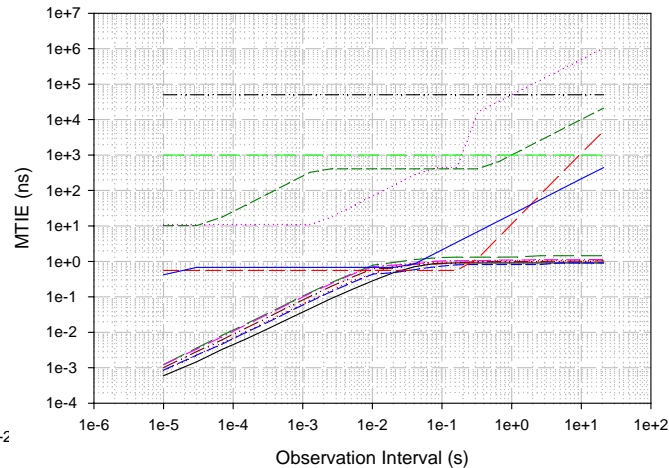
Case 3  
Filtered Phase Adjustments  
No Frequency Adjustments



Case 4  
Instantaneous Phase Adjustments  
Instantaneous Frequency Adjustments



Case 4  
Filtered Phase Adjustments  
Instantaneous Frequency Adjustments



## □ Assumptions

- With clock phase noise (model described in Appendix)
- Granularity of clock = 1 ns
- No frequency adjustments (Case 5); Instantaneous frequency adjustments (Case 6)
  - Inter-message time ( $T_m$ ) = 10 ms; time between frequency offset updates = 100 ms (Case 6)
- Offset between master→slave and slave→master messages initialized randomly at each node
  - All nodes send messages at local free-running clock rate (offsets vary over simulation)

## □ Results (see plots on following slides)

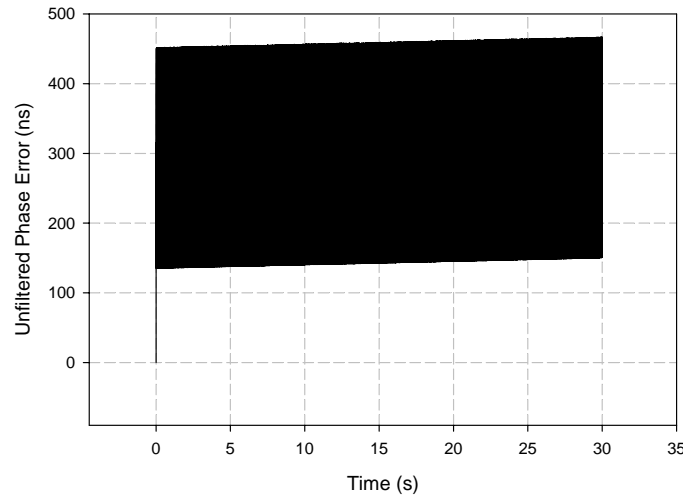
- If frequency adjustments are not made, phase steps occur due to variation in time offset between master→slave and slave →master messages
  - This time offset results in a phase error on the order of the size of the offset (in units of time) multiplied by the fractional frequency difference between the free-running master and slave clocks
  - As the time offset increases from 0 to  $T_m$  (or decreases from  $T_m$  to 0) phase offset changes
  - When the time offset reaches  $T_m$  (or 0) it jumps to 0 (or  $T_m$ ) as one message “walks past” the other
  - This produces a step change in phase error of order  $yT_m$ , where  $y$  is the relative frequency offset between the master and slave
    - E.g., for  $T_m = 0.01$  s and  $y = 100$  ppm, the phase error jump is on the order of 1000 ns

## □ Results (Cont.)

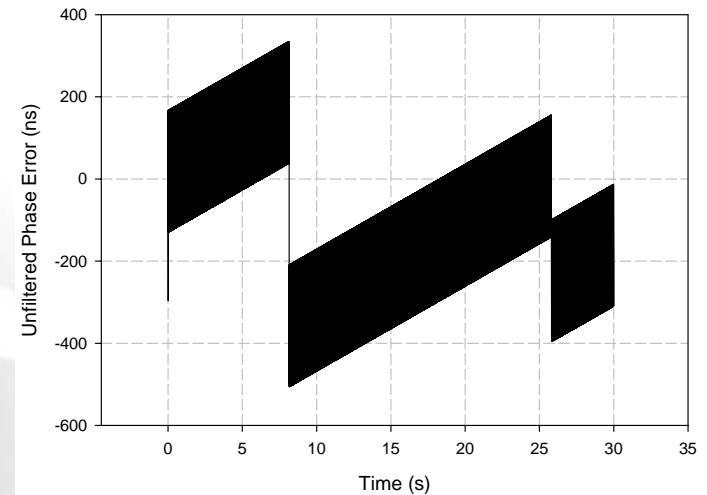
- The 10 Hz filter removes the fast phase variation due to instantaneous phase adjustments, clock phase noise, and non-zero granularity; however, it cannot remove the phase variation due to variation in the time offset between the master→slave and slave→master messages as this variation is much slower
- The effect does not occur when frequency adjustments are made because the error in phase correction due to the frequency offset between the nodes is corrected for
- MTIE for the case with frequency adjustments is roughly the same as in the corresponding case where the master→slave and slave→master message time offset does not vary (Case 4)
- Phase variation does not increase monotonically with number of clocks in chain (in all cases)
- Note that the results exhibit large statistical variability
  - Must run multiple, independent replications of the simulations to obtain confidence intervals for the results

# Simulation Case 5

Case 5, Node 1  
Instantaneous Phase Adjustments  
No Frequency Adjustments

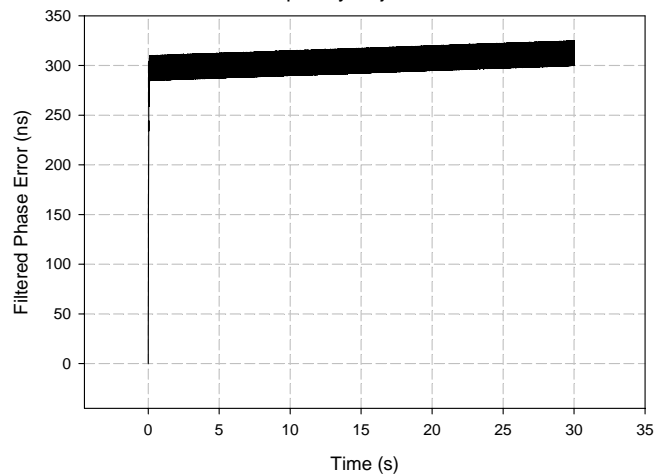


Case 5, Node 10  
Instantaneous Phase Adjustments  
No Frequency Adjustments

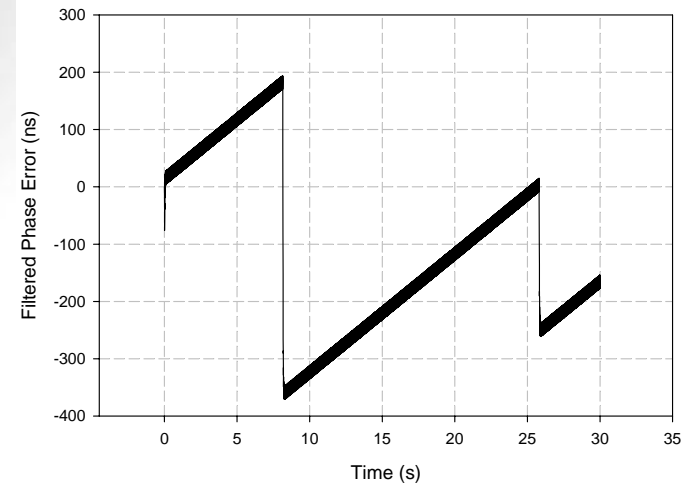


Note:  
Peak-to-peak  
phase variation  
for Case 5 is  
much larger than  
for Cases 3, 4,  
and 6.

Case 5, Node 1  
Filtered Phase Adjustments  
No Frequency Adjustments



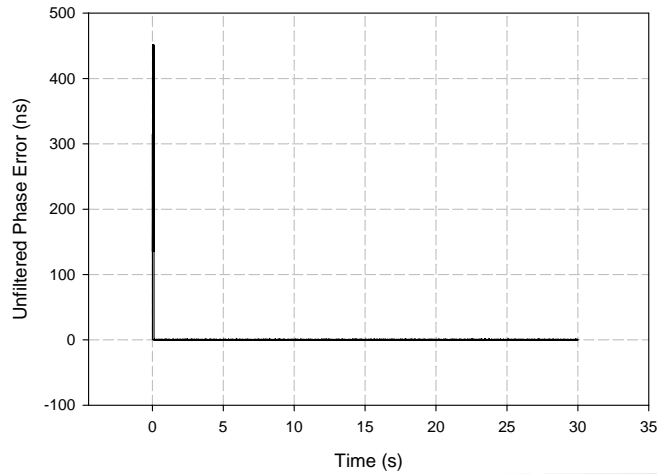
Case 5, Node 10  
Filtered Phase Adjustments  
No Frequency Adjustments



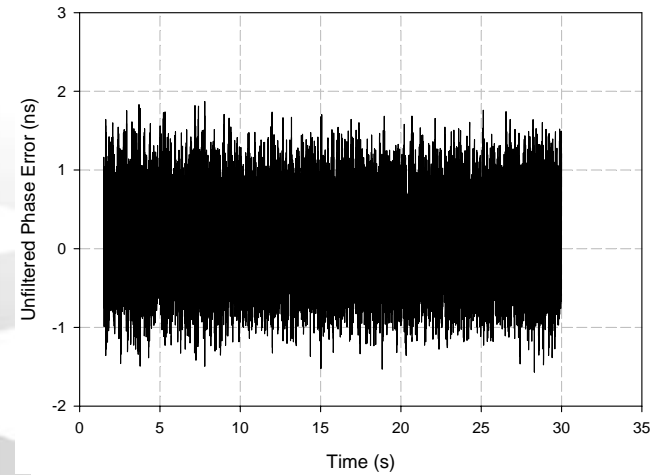


# Simulation Case 6

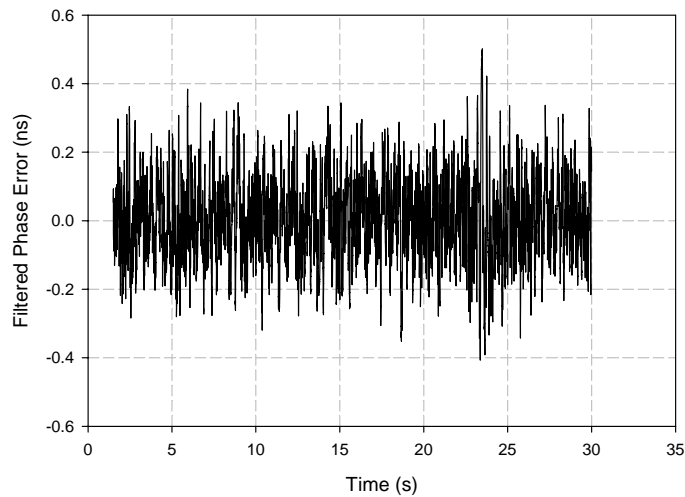
Case 6, Node 1  
Instantaneous Phase Adjustments  
Instantaneous Frequency Adjustments



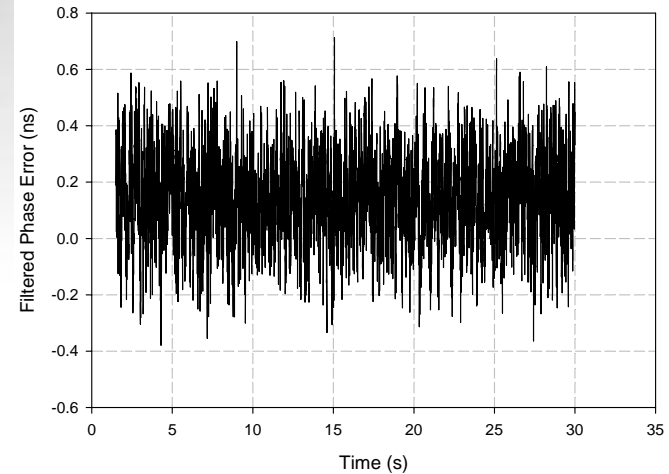
Case 6, Node 10  
Instantaneous Phase Adjustments  
Instantaneous Frequency Adjustments  
(Plot begins after initial transient has decayed)



Case 6, Node 1  
Filtered Phase Adjustments  
Instantaneous Frequency Adjustments  
(plot begins after initial transient has decayed)

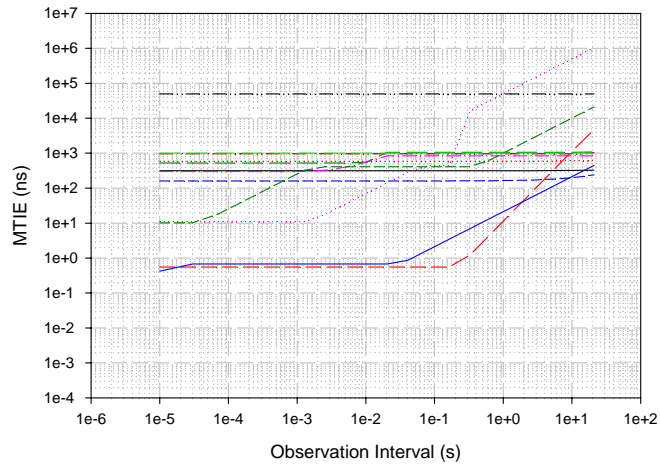


Case 6, Node 10  
Filtered Phase Adjustments  
Instantaneous Frequency Adjustments  
(plot begins after initial transient has decayed)

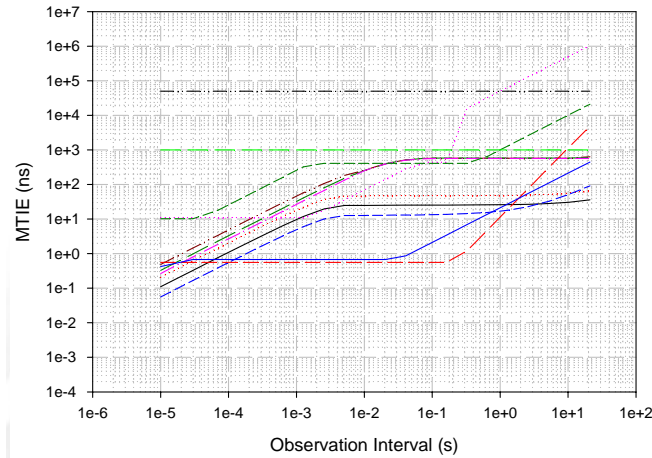


# Simulation Cases 5 and 6

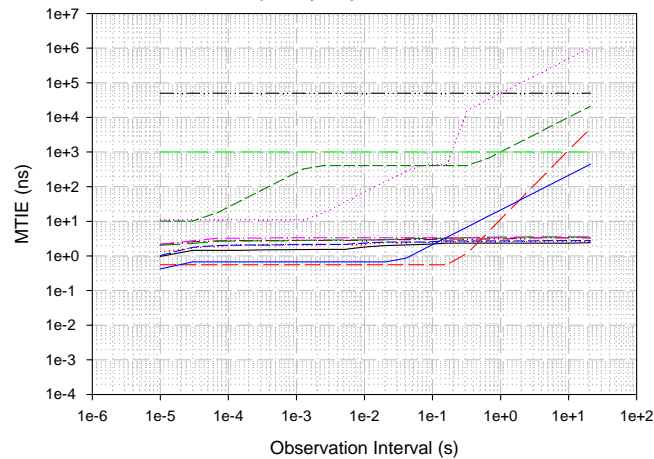
Case 5  
Instantaneous Phase Adjustments  
No Frequency Adjustments



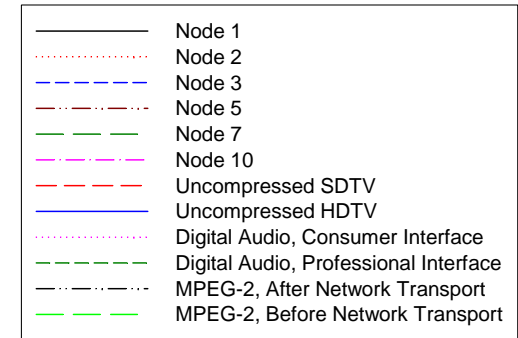
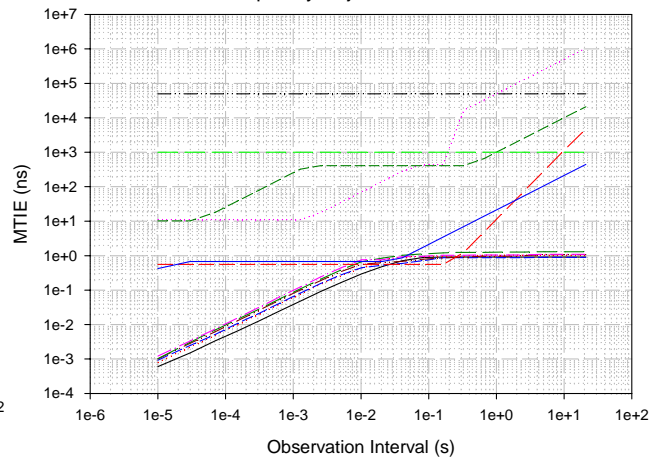
Case 5  
Filtered Phase Adjustments  
No Frequency Adjustments



Case 6  
Instantaneous Phase Adjustments  
Instantaneous Frequency Adjustments



Case 6  
Filtered Phase Adjustments  
Instantaneous Frequency Adjustments



- In ideal case of no clock noise, zero phase granularity, and no variation in the time offset between the master→slave and slave→master messages , can achieve extremely small peak-to-peak phase variation in steady state
  - 0.07 ns with no filtering and frequency adjustments (Case 2, node 10)
  - 0.00055 ns with filtering and frequency adjustments (Case 2, node 10)
  - 0.12 ns with filtering and no frequency adjustments (Case 1, node 10)
- However, with clock noise (using the model of the appendix) and 1 ns phase granularity, peak-to-peak phase variation in steady state is larger
  - 2 – 4 ns with no filtering and frequency adjustments, whether or not time offset between the master→slave and slave→master messages vary
  - 1 – 1.5 ns with filtering and frequency adjustments, whether or not time offset between the master→slave and slave→master messages vary
  - 10 – 50 ns with filtering and no frequency adjustments if time offset between the master→slave and slave→master messages does not vary
  - 35 – 600 ns with filtering and no frequency adjustments if time offset between the master→slave and slave→master messages does vary

- ❑ The cases with clock noise and 1 ns phase granularity indicate that MTIE masks for uncompressed digital video are exceeded if filtering is not done
  - This indicates that filtering is necessary, whether or not instantaneous frequency adjustments are made
  - The end-to-end digital audio masks are met for this case only if frequency adjustments are made
- ❑ The uncompressed digital video masks are slightly exceeded with 10 Hz, 0.1 dB filtering if frequency adjustments are made; they and the consumer interface audio mask are exceeded if frequency adjustments are not made
  - Note that the masks apply to the end-to-end application
    - ResE gets only a budget allocation of the total
    - Get some additional phase variation (likely small) due to the finite granularity of the application time stamps relative to the synchronization signals described here
  - This means it is likely that the filter must have BW that is somewhat narrower than 10 Hz
- ❑ Results show that if instantaneous frequency adjustments are not made, must ensure that master→slave and slave→master messages are sent at nominally the same rate, to avoid variation of their time offset and resulting large phase variation for this case
- ❑ Note that only variations (2) – (4) (see slide 8) have been addressed here

- ❑ Analysis of additional parameter variations
  - Filter BW
  - Time between messages
  - Time between frequency adjustments
  - Larger clock noise level
    - Choose level that bounds noise in oscillators expected to be used in ResE
  - Clock phase granularity
- ❑ Consideration of error in measurement of times the time stamps are sent and received, for implementation of the measurement in different layers
- ❑ Determination of statistical confidence intervals for MTIE (and possibly TDEV) by running multiple, independent replications a simulation case
- ❑ Analysis of other variations/choices (slide 8)

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2. Geoffrey M. Garner, *End-to-End Jitter and Wander Requirements for ResE Applications*, Samsung presentation at May, 2005 IEEE 802.3 ResE SG meeting, Austin, TX, May 16, 2005. Available via [http://www.ieee802.org/3/re\\_study/public/index.html](http://www.ieee802.org/3/re_study/public/index.html).
3. Ralf Steinmetz, *Human Perception of Jitter and Media Synchronization*, IEEE JSAC, Vol. 14, No. 1, January, 1996, pp. 61 – 72.
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5. ITU-T Recommendation G.810, *Definitions and Terminology for Synchronization Networks*, ITU-T, Geneva, August, 1996, Corrigendum 1, November, 2001.
6. ITU-T Recommendation G.8251, *The Control of Jitter and Wander within the Optical Transport Network (OTN)*, ITU-T, Geneva, November, 2001, Amendment 1, June, 2002, Corrigendum 1, June, 2002.
7. Geoffrey Garner, *Jitter Analysis for Asynchronous Mapping of a Client Signal into an OCh*, Lucent Contribution to ITU-T Q 11/15 Interim Meeting, Ottawa, ON, July, 2000.
8. *Phase Noise*, Vectron International, Application Note, available at <http://www.vectron.com>.

9. *Jitter and Signal Noise in Frequency Sources*, Raltron, Application Note, available at <http://www.raltron.com/>
10. David W. Allan, Marc A. Weiss, and James L. Jespersen, *A Frequency Domain View of Time Domain Characterization of Clocks and Time and Frequency Distribution Systems*, Forty-Fifth Annual Symposium on Frequency Control, Los Angeles, CA, May 29 – 31, 1991, pp. 667 – 678.
11. Stefano Bregni, *Synchronization of Digital Telecommunications Networks*, Wiley, 2002.
12. J.A. Barnes and Stephen Jarvis, Jr., *Efficient Numerical and Analog Modeling of Flicker Noise Processes*, National Bureau of Standards, NBS Technical Note 604, June, 1971.
13. James A. Barnes and Charles A. Greenhall, *Large Sample Simulation of Flicker Noise*, 19<sup>th</sup> Annual Precise Time and Time Interval (PTTI) Applications Planning Meeting, December, 1987.
14. Giovanni Corsini and Roberto Saletti, *A  $1/f^\nu$  Power Spectrum Noise Sequence Generator*, IEEE Transactions on Instrumentation and Measurement, Vol. 37, No. 4, December, 1988, pp. 615 – 619.
15. Alexei Beliaev, *Latency Sensitive Application Examples*, Gibson Labs, part of *Residential Ethernet Tutorial*, IEEE 802.3 meeting, March, 2005.

□ Clock phase noise may be modeled as a sum of random processes with power spectral density (PSD) of the form  $Af^{-\alpha}$

▪ In practice, the PSD has 3 terms (see [8] and [9])

- $\alpha = 0$ , White Phase Modulation (WPM)
- $\alpha = 1$ , Flicker Phase Modulation (FPM)
- $\alpha = 3$ , Flicker Frequency Modulation (FFM)

▪ Can write the PSD,  $S_x(f)$  as

$$S_x(f) = \frac{A}{f^3} + \frac{B}{f} + C, \text{ where } S_x(f) \text{ has units of ns}^2/\text{Hz}$$

▪ Often express as

$$S_\phi(f) = (2\pi\nu_0)^2 S_x(f), \text{ where units of } S_\phi(f) \text{ are rad}^2/\text{Hz}$$

▪ An example PSD specification is given in Figure 12 of [8], and reproduced on the next slide

- Data in [8] is given in dBc/Hz; data has been converted to rad<sup>2</sup>/Hz
- Data in [8] is given only for frequencies below 10 kHz; here, we assume the PSD is flat above 10 kHz
- Dotted curve on the next slide is the converted data of [8]; solid line is a conservative fit of the above power law sum

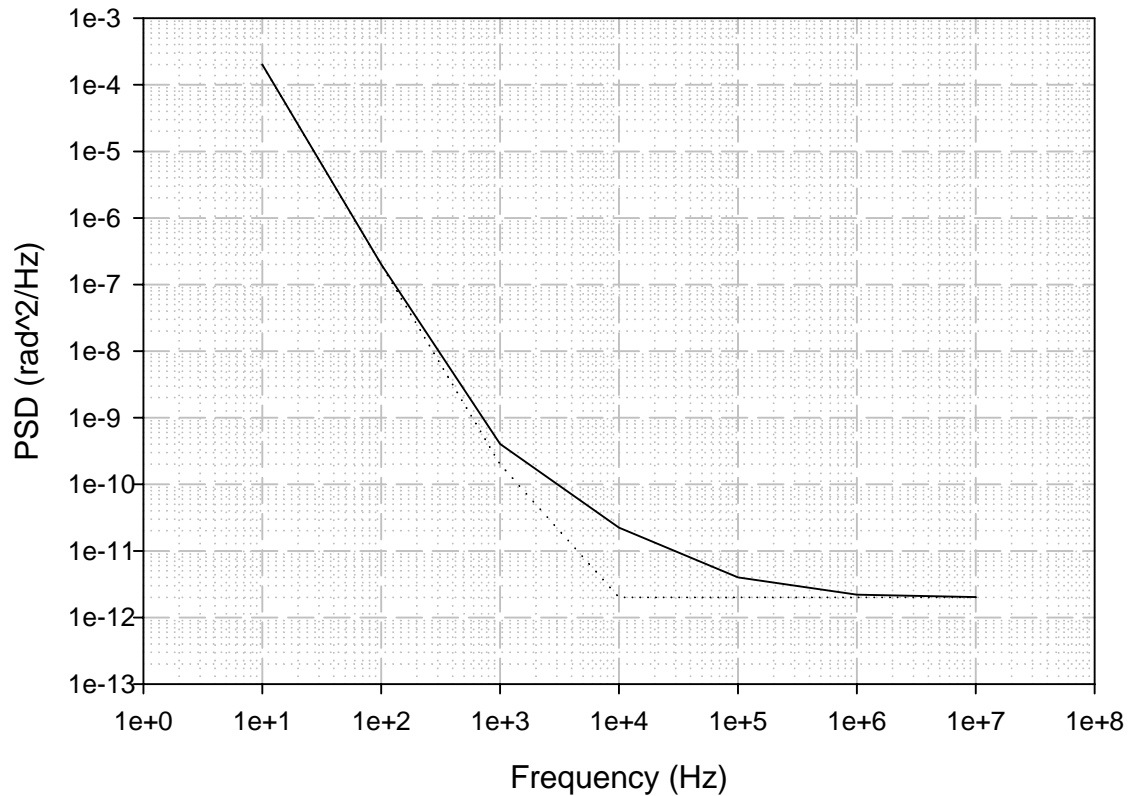
▪ The specifications for the individual products of [7] and [8] are below this example, at least for those products where phase noise specifications are provided



# Appendix I - Clock Noise Model

**Example Clock Phase Noise Specification Provided in [9] (data in [9] does not extend above 10 kHz; PSD is assumed flat for higher frequencies with the 10 kHz value)**

— analytic form of PSD  
..... specification in [9]



Note: Data in [8] is given in dBc/Hz; data has been converted to rad<sup>2</sup>/Hz

- Another measure for clock noise, which is more convenient because it is a time domain parameter, is Time Variance (TVAR)
  - Time Deviation (TDEV) is the square root of TVAR
- TVAR is 1/6 times the expectation of the square of the second difference of the phase error averaged over an interval

$$\text{TVAR}(\tau) = \frac{1}{6} E\left[\left(\Delta^2 \bar{x}\right)^2\right]$$

where  $E[\cdot]$  denotes expectation,

$\bar{x}$  denotes average over the integration time  $\tau$ ,

and  $\Delta^2$  denotes second difference

- TVAR may be estimated from measured or simulated data using [5]

$$\text{TVAR}(n\tau_0) = \frac{1}{6n^2(N-3n+1)} \sum_{j=1}^{N-3n+1} \left[ \sum_{i=j}^{n+j-1} (x_{i+2n} - 2x_{i+n} + x_i) \right]^2, \quad n = 1, 2, \dots, \text{integer part}(N/3)$$

where  $\tau_0$  is the sampling interval and  $\tau = n\tau_0$

- TVAR is equal to  $\tau^2/3$  multiplied by the Modified Allan Variance
- For power-law noises with PSD proportional to  $f^{-\alpha}$ , TVAR is proportional to  $\tau^\beta$ , where  $\beta = \alpha - 1$
- The magnitude of TVAR may be related to the magnitude of PSD for power-law noises; see [10] and [11] for details

- FFM

$$S_x(f) = \frac{A}{f^3} \quad \text{TVAR}(\tau) = \frac{(2\pi)^2 9 \ln 2}{20} A \tau^2$$

- FPM (result is from [10]; a more exact expression is given in [11])

$$S_x(f) = \frac{B}{f} \quad \text{TVAR}(\tau) = \frac{3.37}{3} B$$

- WPM

$$S_x(f) = C \quad \text{TVAR}(\tau) = \frac{\tau_0 f_h}{\tau} C$$

$f_h$  = noise bandwidth

## □ Simulation of WPM

- WPM is simulated as a sequence of independent, identically distributed random samples
- Noise distribution is taken as Gaussian with zero mean
- Variance and sampling time determine TDEV level
  - Choose variance such that, with given sampling time, the computed TDEV from a sample history is close to value obtained from above relation between TDEV and PSD
    - Assume noise bandwidth is equal to line rate (100 MHz)

## □ Simulation of FPM

- FPM is simulated by passing a sequence of independent, identically distributed random samples through a Barnes/Jarvis filter [12] – [14]
  - If white noise is input to a filter with frequency response  $H(f) = f^{-1/2}$ , the output is a random process with PSD proportional to  $1/f$
  - The Barnes/Jarvis filter approximates an  $f^{-1/2}$  frequency response using a bank of lead/lag filters
    - The actual frequency response of this filter is a “staircase”
    - The spacings of the poles and zeros are chosen such that the average slope is  $-10$  dB/decade

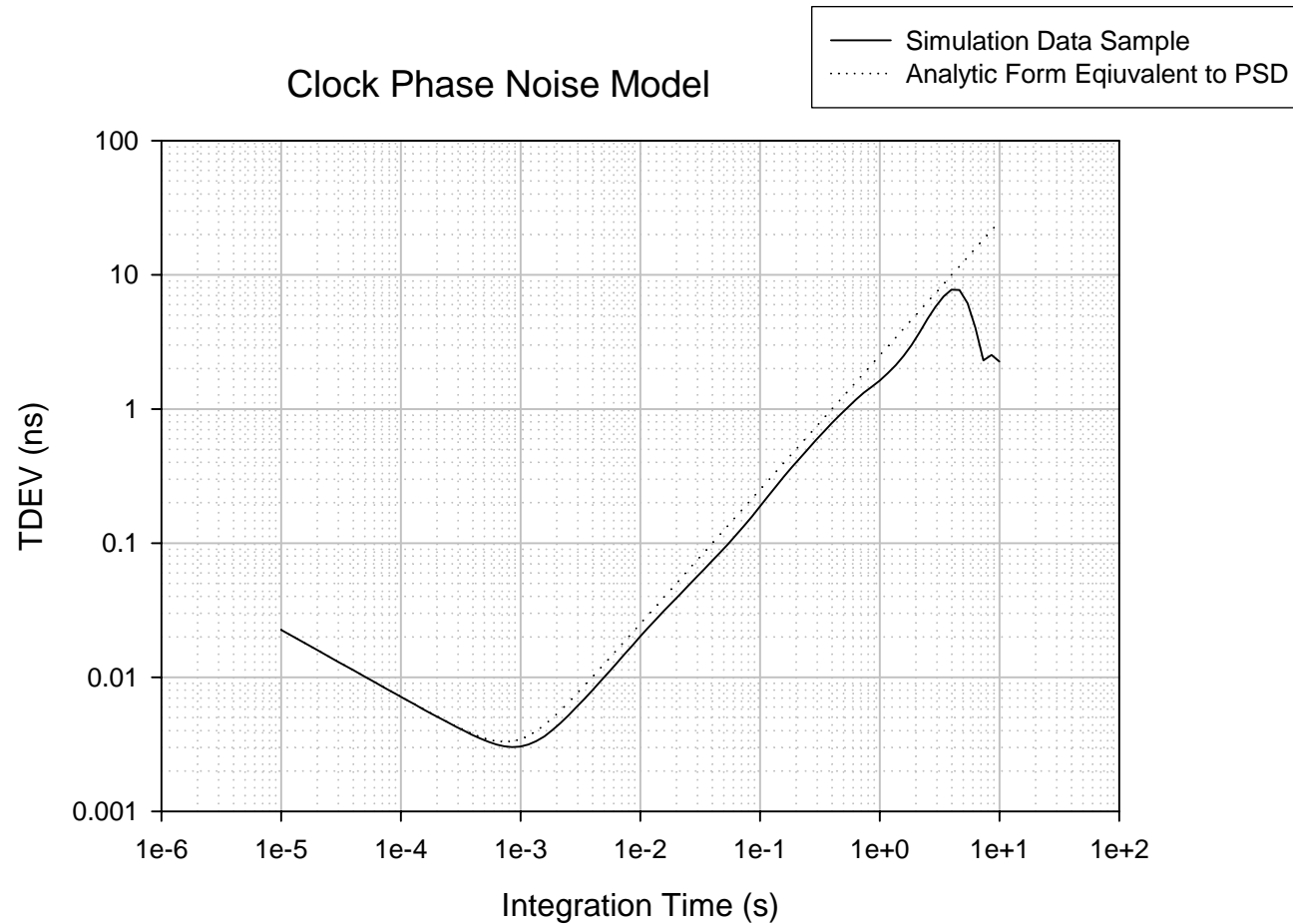
## □ Simulation of FPM (Cont.)

- Noise distribution is taken as Gaussian with zero mean
- Variance determines TDEV level
  - Choose variance such that the computed TDEV from a sample history is close to value obtained from above relation between TDEV and PSD

## □ Simulation of FFM

- Input a sequence of independent, identically distributed random samples through a Barnes/Jarvis filter followed by an integrator (accumulator)
- Noise distribution is taken as Gaussian with zero mean
- Variance determines TDEV level
  - Choose variance such that the computed TDEV from a sample history is close to value obtained from above relation between TDEV and PSD

□ Next slide shows TDEV for simulated data sample ( $10^{-5}$  s time step) and analytic form equivalent to PSD (solid curve on slide 35)



- Jitter and wander requirements can be expressed in terms of Maximum Time Interval Error (MTIE) masks
- MTIE is peak-to-peak phase variation for a specified observation interval, expressed as a function of the observation interval
  - An estimate of MTIE may be computed by (see [5])

$$\text{MTIE}(n\tau_0) \cong \max_{1 \leq k \leq N-n} \left( \max_{k \leq i \leq k+n} x(i) - \min_{k \leq i \leq k+n} x(i) \right), \quad n = 1, 2, \dots, N-1$$

where  $\tau_0$  is the sampling interval,  $n\tau_0$  is the observation interval,  $x(i)$  is the  $i^{\text{th}}$  phase sample, and  $N$  is the number of phase samples ( $N\tau_0$  is the measurement interval)

- The derivation of the MTIE masks on slide 6 from the jitter and wander requirements is given in [2]