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A distributed approach to topology discovery in grid environment

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1 The Grid

Today **Grid Computing** is ubiquitous. As of June 2005 searching those 2 words on *Google* would return more than 13 millions results and just the first ten results show how some of the largest technology companies in the world, like *IBM*, *Intel*, *Sun Microsystems* and *HP* have decided to invest in their own **Grid Computing** project.

Grid is a type of parallel and distributed system that enables the sharing, selection, and aggregation of geographically distributed *autonomous* resources dynamically at runtime depending on their availability, capability, performance, cost, and users' quality-of-service requirements.

The key distinction between clusters and grids is mainly lie in the way resources are managed. In case of clusters, the resource allocation is performed by a centralised resource manager and all nodes cooperatively work together as a single unified resource. In case of Grids, each node has its own resource manager and doesn't aim for providing a single system view. Another important distinction is that in cluster environments all the applications running are cluster-oriented while every host of a Grid is independent from others and runs many applications non grid-aware. More information on **Grid Computing** concepts can be retrieved here [2],[3],[4]

According to *Wikipedia* [1] **Grid Computing** offers a model for solving massive computational problems by making use of the unused resources (CPU cycles and/or disk storage) of large numbers of disparate, often desktop, computers treated as a virtual cluster embedded in a distributed telecommunications infrastructure. **Grid Computing**'s focus on the ability to support computation across administrative domains sets it apart from traditional computer clusters or traditional distributed computing.

Grids offer a way to solve Grand Challenge problems like protein folding, financial modelling, earthquake simulation, climate/weather modelling etc. Grids offer a way of using the information technology resources optimally in an organization. Some of the large companies involved in Grid development, such as *IBM* [34], provide a means for offering information technology as a utility bureau for commercial and non-commercial clients, with those clients paying only for what they use, as with electricity or water.

Grid Computing has the design goal of solving problems too big for any single super-computer, whilst retaining the flexibility to work on multiple smaller problems. Thus grid computing provides a multi-user environment. Its secondary aims are: better exploitation of the available computing power, and catering for the intermittent demands of large computational exercises.

Just to sketch the idea of what **Grid Computing** really turns to be in the real world here

are reported three interesting definition of **Grid Computing** from Wikipedia:

- CERN, who were key in the creation of the World Wide Web, talk of The Grid: *a service for sharing computer power and data storage capacity over the Internet*.
- Pragmatically, **Grid Computing** is attractive to geographically-distributed non-profit collaborative research efforts like the NCSA Bioinformatics Grids such as BIRN: external grids.
- **Grid Computing** is also attractive to large commercial enterprises with complex computation problems who aim to fully exploit their internal computing power: internal grids.

1.1 GGF, the Global Grid Forum and GTx, the Globus Toolkit

The Global Grid Forum (GGF) has the purpose of defining specifications for **Grid Computing**. The Globus Alliance implements these standards through the **Globus Toolkit**, which has become the de facto standard for grid middleware¹. As a middleware component, it provides a standard platform for services to build upon, but **Grid Computing** needs other components as well, and many other tools operate to support a successful Grid environment. This situation resembles that of TCP/IP: the usefulness of the Internet emerged both from the success of TCP/IP and the establishment of applications such as newsgroups and webpages.

Globus has implementations of the GGF-defined protocols to provide:

- Resource management: Grid Resource Allocation & Management Protocol (GRAM)
- Information Services: Monitoring and Discovery Service (MDS)
- Security Services: Grid Security Infrastructure (GSI)
- Data Movement and Management: Global Access to Secondary Storage (GASS) and GridFTP

A number of tools function along with Globus to make **Grid Computing** a more robust platform, useful to high-performance computing communities. They include:

- Grid Portal Software such as GridPort and OGCE

¹ Software that connects two otherwise separate applications in case belonging to different computing framework such as lying on different operative systems or belonging to different network domains.

- Grid Packaging Toolkit (GPT)
- MPICH-G2 (Grid Enabled MPI)
- Network Weather Service (NWS) (Quality-of-Service monitoring and statistics)
- Condor (CPU Cycle Scavenging) and Condor-G (Job Submission)

Most of the grids which span research and academic communities in North America and Europe use the `Globus Toolkit` as their core middleware.

`XML`-based web services offer a way to access the diverse services/applications in a distributed environment. In the last few years the worlds of grid computing and of web services have started to converge to offer Grid as a web service (Grid Service). The Open Grid Services Architecture (`OGSA`) has defined this environment, which will offer several functionalities adhering to the semantics of the Grid Service.

`OGSA`, `OGSI` and their implementation provided by `Globus Toolkit` will be explored into the details later [B.0.4](#)

2 Network awareness and Topology discovery in grid environments

A difficult problem in designing efficient applications for computational Grids is that the wide-area interconnection network is highly complex, dynamic, and heterogeneous. A more traditional computing platform such as a cluster or supercomputer typically has a highly regular network with constant, guaranteed performance. With a Grid, however, the network has an irregular and asymmetric topology, where different links have different speeds, which even change over time.

On the other hand, the dynamic joint optimization of both computational and network resources has the potential of guaranteeing optimal performance to Metropolitan Area Network (MAN) grid applications. For this purpose the introduction of grid network services has been proposed. Grid network services provide either the users or the programming environment, with information on network resource status and means for reallocating network resources.

2.1 The goal of this work

In this study a specific grid network service, i.e. the topology discovery service, is presented. The topology discovery service provides grid users with up to date information on the

grid network status considering different network layers. The proposed implementation is based on a producer/consumer architecture where grid users are contemporarily producers and consumers of the required information. This implementation can be utilized in any MAN based on commercial routers without the need of modifying router management and control protocols. The information provided by the proposed system could be useful to allow applications to handle with the network taking into consideration the link level thus improving the effectiveness at which operations are performed. For instance, a grid user application, empowered by network awareness, could decide to route a connection to another application via an alternative path instead of the default provided by the IP level route because of congestion. Other examples on the importance of exploiting topology-related information in Grid environments will be given in the next sections.

2.2 Performance measures over grid networks

The GGF Network Measurements Working Group has provided a draft [8] in which is described a set of network characteristics and a classification hierarchy for these characteristics useful for Grid applications and services.

The goal of their work is to identify the various types of network measurements according to the network characteristic they measure and the network entity on which they are taken in the hope that the application of this naming system will facilitate the creation of common schemata for describing network monitoring data in Grid Monitoring and Discovery Services.

The NMWG is closely related to the IETF Internet Protocol Performance Metrics (IPPM) WG.

Whereas their work focuses on best-practices for network engineers and defining the correct way to take measurements, the GGF NMWG's goal is creating a comprehensive system that can be used to categorize all measurements that are in use, taking into account the requirements of grid applications and grid-related tools.

2.2.1 Sample Grid use of network measurements

As an example of how network measurements could be used in a Grid environment, the case of a Grid file transfer service is presented. Assume that a Grid Scheduler [10] determines that a copy of a given file needs to be copied to site A before a job can be run. Several copies of this file are registered in a Data Grid Replica Catalogue [11], so there is a choice of where to copy the file from. The Grid Scheduler needs to determine the optimal method to create this new file copy, and to estimate how long this file creation will take. To make

this selection the scheduler must have the ability to answer these questions:

1. what is the best source (or sources) to copy the data from?
2. should parallel streams be used, and if so, how many?
3. what TCP window and buffer size should be used?

Selecting the best source to copy the data from requires a prediction of future end-to-end path characteristics between the destination and each possible source. However also an on-the-fly measurement of the performance obtainable from each source in terms of available bandwidth (both end-to-end and hop-by-hop), latency, loss, and other characteristics could be helpful to improve file transfer performance.

Determining whether there would be an advantage in splitting up the copy, and, for example, copying the first half of the file from site B and in parallel copying the second half of the file from site C, requires hop-by-hop link availability information for each network path. If the bottleneck hop is a hop that is shared by both paths then there is no advantage to splitting up the file copy in this way. Parallel data streams will usually increase the total throughput on uncongested paths, as shown by Hacker et al. [12]. However, on congested hops, using parallel streams may just make the problem worse. Therefore, further measurements, such as delay and loss, are needed to determine how many parallel streams to use.

Even in the case of a single stream, accurate network measurements can be used to greatly improve performance and host resource allocation. TCP has always used a “prediction” (or smoothed estimate) of the RTT to determine timeouts. Recent work in the Web100 [13] project aims to improve these estimates and make more information available to the TCP stack and to external programs.

2.2.2 Terminology

After sketching some ideas about how network measurements can be successfully exploited in a **Grid Computing** environment a more precise definition of *network topology* and *performance measurement* is needed.

As indicated in figure (1), a network topology is composed of Nodes and Network Paths. Each Network Entity is annotated with attributes including protocol and QoS level. Network Paths are used to represent the connection between any two nodes in the network. Network Paths can represent anything from an end-host to end-host connection across the Internet to a single link between two Ethernet switches. Representing a network’s topology

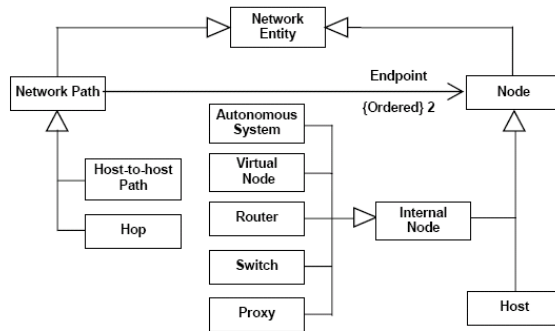


Figure 1: The relation between the node and path Network Entities. The internal nodes shown are only examples, and are not exclusive

with link-layer paths allows a detailed model of a network's structure to be built. Multicast nor broadcast issues and focuses on point-to-point paths are not considered

2.2.3 Representing Topology

Network observations can be reported for a variety of different types of network entities:

- **End-to-End Paths** are the common case of host-to-host measurements.
- **Links** between routers or switches are frequently measured for capacity, availability, latency and loss.
- **Nodes** may report useful information such as router queue discipline or host interface speed.

For systems that report network topology - the network entities themselves: hosts, routers, switches, and links - form the actual observations being reported rather than just the object being observed. Topologies are best representing using a graph structure, with nodes and edges.

Furthermore, different systems care about topology at different levels. An ISP or network manager focusing on SLA satisfaction might be most interested in a topology of AS to AS, reflecting the different ISP's contractual relationships. Many systems collect topology at layer-3 through traceroute, typically discovering all the routers along a path. Other systems might report only layer-2 topology between routers, or both layer-2 and layer-3 topology. Most often, a system does not have access to the layer-2 devices across an entire end-to-end path, so it might report the layer-2 topology at one or both leaf networks, but be unable to discover more than the layer-3 routers across the network core.

2.2.4 Physical and Functional Topologies

There are two approaches to characterizing network topology: physical and functional. The physical approach determines the physical hops that connect the network together. By determining the connections between nodes, along with their capacities, queuing disciplines, and traffic load, the network can be modelled and its behaviour analysed or predicted. Physical topology can be determined for both LANs and WANs.

The functional approach differs in that it makes use of end-to-end information, under the assumption that such observations are more readily available and usable than modelling low level network behaviour. Functional topology representations attempt to group and arrange network sites according to their perceived closeness determined by traffic performance, rather than according to the actual connections of physical hops. This approach may be taken across a variety of sites distributed around the Internet, or using a single source tree. Functional topologies provide useful information that an application could use to make many of the same decisions that are made with physical information. The only noticeable difference in the representation of physical and functional topologies is, as it was mentioned before, that functional topology cannot handle with shared hops simply because it doesn't perceive it.

Nodes are generally classified into hosts and internal nodes:

- Hosts are considered to be only endpoints of communication. Internal nodes are capable of forwarding traffic, and can be routers, switches, or proxies, as well as more general concepts such as autonomous systems or virtual nodes in a topology.
- Virtual nodes are used to describe additional functionality that might be found in a physical node. Note that physical and virtual nodes do not occupy disjoint graphs in a network topology. In particular, functional networks always contain physical nodes at their edges. There may be several virtual nodes, describing different functionality, overlaid on a single physical node. Furthermore, virtual nodes may play a role in describing physical networks.

Two examples of the use of a virtual node are given:

- Because some hosts also perform routing, a virtual node might be used to represent the routing done by the host, with the host being a separate node linked to the virtual node.
- As an example of using a virtual node to describe a physical network, consider a halfduplex link such as a wireless network. Because paths are defined to be unidirectional,

tional, the half-duplex nature of such a network cannot be adequately described in a path. However, by imposing a virtual node in the network with appropriate characteristics of the half-duplex link, the virtual node can represent the transmission characteristics of the half-duplex link.

Like paths, nodes also should be annotated with a set of attributes. The attributes indicate a specific set of characteristics of the node when handling a particular type of traffic. In particular, queuing disciplines, queue lengths, etc. may vary quite significantly according to type of traffic and QoS level.

2.2.5 An alternative model for topology description

In their paper [27] Lacour et al. propose a description model of (grid) networks which provides a functional view of the network topology slightly different from the ones proposed before.

The grid network description's aim is to be used to select node before mapping an application to the resources of a grid. This goal can be achieved by representing a logical network topology and by **grouping** together the computers with common network characteristics. That amounts to not representing all the physical network connections. For instance, a Internet connection can be modeled using just one logical link, while multiple physical paths may interconnect different Internet domains.

Lacour et al. assert that to serve the purpose of application deployment, the network topology description does not need to be aware of any switch or router, neither does it need to represent every single network link. This assertion contradicts the assumptions made by other [6] who claims that representing shared network links is essential for a good description of grid network topology. Indeed, the effect of shared links is that communication performance can decrease over certain network paths at certain time periods: rather than including this effect in the network topology the effect of shared links can be taken into consideration in the numerical network properties, such as jitter, bandwidth variance (in time), etc.

All the grid network topology description needs to include is the fact that a certain set of computers are connected together over the same sub-network, and that those computers have roughly the same communication characteristics while communicating with each other. So the computers are registered to network groups, depending on how many sub-networks they belong to. Thus, the common communication capabilities (end-to-end bandwidth, latency) of the computers belonging to the same sub-network are entered only once as attributes of the network group, as well as the software available to access particular

network technologies (BIP, GM or MX for a Myrinet network, for instance).

Network grouping makes the node selection phase easier because it supplies a functional view of network topology highly hierarchical. Intuitively, grouping replaces a “join” (in the database language) for free, since the information about the network is already “precompiled”. As each individual link could not be taken into consideration the description becomes more compact. Lacour et Al. also assert that network grouping makes sense because end-to-end network performance properties are roughly the same between any two computers of a sub-network like a dedicated cluster or a LAN: network performance characteristics can be described as attributes of network groups using the results of the NMWG from the GGF.

The topology described using the network grouping approach uses a directed acyclic graph (DAG). The nodes of the grid network description graph correspond to the network groups or the computers (which can be considered as network groups made of just one host).

The oriented edges of this description graph correspond to network group inclusions: network groups can have parent or child network groups and the edges are oriented from a parent network group to a child network group. In other words, a child network group represents a sub-network of its parent network group.

2.3 Common performance metrics

The common network performance indexes of more interest are:

- **bandwidth capacity:** the theoretical maximum link-layer bandwidth of a network element or end-to-end path.
- **bandwidth utilization:** the aggregate of all traffic currently being consumed on a hop or path.
- **available bandwidth:** the capacity minus utilization over a given time interval.
- **achievable bandwidth:** the throughput between two end points given a specific set of conditions, such as transmission protocol (e.g.: TCP or UDP), host hardware (e.g.: processor speed, bus speed, NIC speed), operating system, system settings (e.g.: TCP buffer size, txqueuelen setting), and so on.
- **delay:** one way delay (OWD) is the time it takes for a packet to travel from source to destination. Round Trip Time (RTT) is the time to travel from the source to the destination and back.

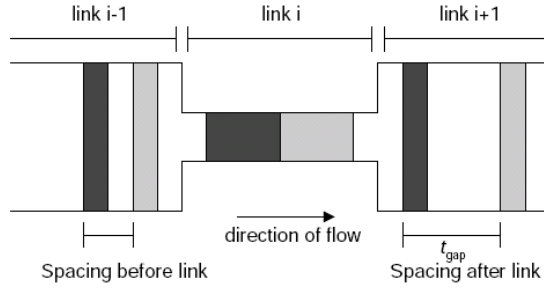


Figure 2: The packet pair technique

- **loss**: fraction of packets lost between network element A and B. Loss can also be one way or round trip.

2.4 Measurement methodologies

Here we take into consideration the common techniques used by network sensors to perform measurements. More details can be found in [9]

2.4.1 Packet pair

With this technique the bandwidth is calculated sending consecutive pairs of packet spaced by some know time.

If non competitive traffic on the path is present, packets will be queued one by one on the slowest link (bottleneck) Comparing the gaps between packets after the link with the initial temporal gap it is possible to estimate the bandwidth under the hypothesis that the delay is proportional to the packets dimension. Capacity is therefore measured in this way:

$$Bandwidth = \frac{P}{gap} = \frac{[bites]}{[secs]}$$

where P is the packet dimension and gap is the temporal spacing between packets. (Figure 2)

2.4.2 Packet train

Packet train is a generalization of the packet pair technique. With this methods an homogeneous ensemble of packets is sent, instead of pairs. The interarrival times are evaluated between pairs belonging to the same ensemble.

2.4.3 Tailgating

This technique aims to improve the packet pair method. It is divided in two phases: during the first phase (σ phase) the performance along an end-to-end path is measured while during the second one *tailgating* phase link-based measures are taken. During the σ phase latency and bandwidth are calculated for the link prior to the destination. The calculation is performed sending a train of packets of different dimensions until the estimate of the bandwidth is sufficiently precise.

In the second phase two packets are sent. The first is as long as the link MTU and has TTL equal to the distance of the link is under measurement. The second packet has the minimum dimension allowed by the protocol. As long as the TTL of the first packet isn't expired the second (smaller) packet follows the first. When the TTL of the first packet expires the second one proceeds without being queued to the first one anymore. This mechanism is iterated for each link to the destination.

In this way it is possible to estimate the bandwidth of intermediate link considering the delay accumulated by the smaller packet compared to the (fixed) delay of bigger packets.

2.4.4 Variable Packet Size (VPS)

With this methodology it is possible to estimate the bandwidth of the link. Packets of different dimension are sent through the network using the TTL field of IP. In this way, after passing a know number of links the packet is discarded and an ICMP message is sent back to the sender so that the sender will be able to calculate the round trip time.

Repeating this mechanism many time and varying the packets dimension it is possible to build a XY graph of the delay vs packet size.

For each dimension is then considered the minimum delay obtained supposing the packet never been queued that is, the delay is only dependant from the transit time through the net.

Then this points are linearly interpolated, so that the difference between the slope of two consecutive nodes is the inverse of the bandwidth of the link which connects those two nodes.

2.4.5 Even/Odd

Even/odd is a mathematic tool to increase the accuracy in the measurements of VPS-like techniques. For each packet dimension the ensemble of obtained samples is divided in two subset: the subset of odd dimension samples and the subset of even dimension samples.

The bandwidth is therefore estimated independently:

1. evaluating only the even-indexed samples
2. evaluating only the odd-indexed samples
3. evaluating only the even-indexed samples through a known index i and the odd-indexed through the end
4. evaluating only the odd-indexed samples through a known index i and the even-indexed through the end

Considering the minimum and the maximum estimate obtained a confidence interval is drawn. The real value will be contained in that interval.

2.4.6 Self Loading Periodic Streams (SLOPS)

With this technique many packet streams are sent at a frequency greater than the supposed bandwidth of the link. The packets will be queued on the node that processes them at a lower frequency. Each packet is given a timestamp when it is sent. Upon the arrival the timestamp of consecutive packets are compared with the interarrival time taking into account the network latency. In order to modulate the frequency to which packet are sent an adaptive algorithm is then used.

2.4.7 TCP simulation

A TCP connection is simulated by means of UDP or ICMP packets. The TCP slow-start phase is simulated as well in order to determine the link MTU

2.4.8 Path flooding

A measurement is performed trying to fill up the link with a flood of packets. In this way the real bandwidth at the applicative level is measured, taking into account all the overheads caused by lower level protocol, operative system and the traffic in transit on the link. This measurement technique is highly intrusive and affects heavily the communications through the measured link.

2.5 Network sensors

Table 2.5 contains mapping of some common network measurement tools to the network characteristics as described in [8]. Table 2.5 table is focused on active monitoring tools, however there are a number of passive tools that can also be used to measure some of the NMWG defined characteristics.

Tool name	Characteristics Measured	Measurement Methodology	Path or hop
bing	bandwidth capacity loss delay (round trip)	variable packet size	path
bprobe	bandwidth capacity	packet pair	path
cprobe	bandwidth utilization	packet pair	path
clink	bandwidth capacity delay (round trip)	variable packet size	hop
Iperf	achievable bandwidth	path flooding	path
netest	available bandwidth achievable bandwidth loss delay (round trip)	packet train and path flooding	path
Netperf	achievable bandwidth	path flooding	path
Nettimer	bandwidth capacity	packet pair with tailgating	path
owping	one-way delay loss reordering	active, GPS-based	path
pathchar	bandwidth capacity loss delay (round trip)	variable packet size	hop
pathload	available bandwidth	SLOPS	path
pipechar	available bandwidth loss delay (round trip)	packet train	hop

ping	delay (round trip) loss	ICMP echo	path
sprobe	bandwidth capacity	packet pair	path
traceroute	topology delay (round trip?)	varied TTL	hop
TReno	available bandwidth	TCP simulation	path
ttcp	achievable bandwidth	path flooding	path
pathrate	bandwidth capacity	packet train	path
pchar	bandwidth capacity loss delay (round trip)	variable packet size	hop

Extensive documentation about network measurement tools can be found here [\[14\]](#) [\[9\]](#), [\[15\]](#)

2.6 pathchar

A comparative analysis and test of the network sensor presented in the previous section has been taken by Rody Schoonderwoerd [\[9\]](#)

Even though his study doesn't take into consideration the tool `pathchar` that at the very end is the one here utilized, Rody draw some basics outlines useful to compare and choose among the plenty of sensors available.

The choice of `pathchar` has been driven by two basic requirements of our infrastructure (that will be explained into the details later)

1. The network sensor have not to be too intrusive since all the measures are taken in the same time. For instance, network sensors that use the path flooding technique cannot be taken into consideration.
2. The tool is to perform a measurement at hop level. Bearing this consideration in mind it is possible to discard all the tools that work on a end-to-end basis
3. The tool should require low level administrative privileges on the host it is installed. `pathchar` needs only to be `suid root` which is the minimum level required for a tool that have to write IP packets directly rather than using the socket interface. This should not considered at all an heavy constraint since the well known and ubiquitous `traceroute` needs to be `suid root` as well. Without that ability, it could not set the time-to-live on the packet and thus wouldn't work.

`pathchar` fits all previously listed requirements. As a drawback, as it will be pointed out, it does not deal very well with high throughput network. Moreover it seems that it also lacks software maintenance and this could mean that no further development will affect it. A. Downey in his work on `pathchar` [35] provides an extensive analysis and testing of the `pathchar` tool pointing out advantages and shortcomings in different network layouts. It is necessary to stress that the network sensor used, even though it can affect greatly the results of the work of the proposed system, is really not the core point. The proposed infrastructure has been thought to be modular thus uncoupled from the low level sensors it uses. With some minor software rewriting the topology discovery service here presented can adapt to using some other low level network sensor.

2.7 Grid Network Services

So far it has been discussed about lower level sensor in charge of performing measurements on the grid network. In the following section the higher level software monitor architecture deemed to elaborate the information provided by sensors and dispatch it to the grid user as a service is taken into consideration.

An informational draft that explains Grid Network Services getting into the details and by means of a plenty of examples and useful implementation directives can be found here [7] Grid network services combine several existing network services yielding a rich mediation function between grid applications and legacy networks. Through these services, the network resource is seen joining CPU and storage as a first-class, grid managed resource (and handled, as such, by a community scheduler, or other OGSA services).

A network service is further labeled as a Grid Network Service whenever the service has roles and/or interfaces that are deemed to be specific infrastructure.

2.7.1 Overview of Grid Network Services

Network services assist a grid infrastructure in different ways. In the simplest setup, a grid application (or a grid infrastructure on its behalf) consults a Network Service as if it was an omniscient oracle (e.g., a directory service) using a plain question/answer style of interaction. In more complex setups, Network Services interact with one another to realize one or more end-to-end feedback loop cycles (as in: observe + request + provision + act). Application requirements, policy considerations, and brokers directives are continuously injected into these feedback loops via expressive languages and machine interfaces (as opposed to, say, point-and-click sessions driven by operators).

Figure (3) shows an example of notional network services engaged in a fairly complex set

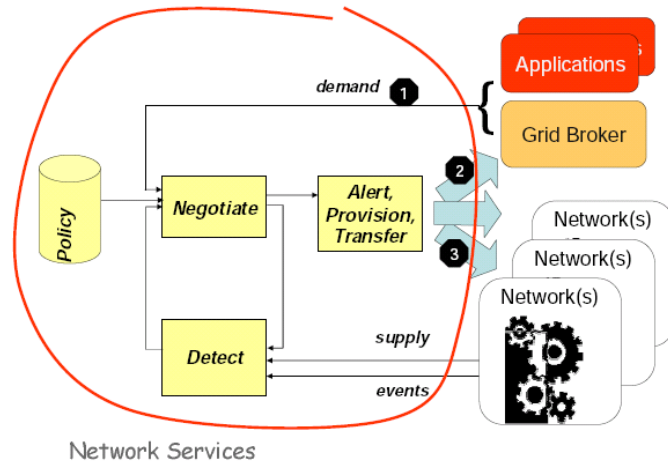


Figure 3: An example of Network Service in action (e.g. for an hypothetical bulk transfer mover)

of feedback loops. Applications demands, policy, and networks observed metrics (capacity, latency, jitter, loss, etc.) are continuously mediated, resulting in data marshalling and provisioning actions upon the network and/o the end-systems.

The various flows defined by boxes and edges must operate in a secure fashion across 1..N administrative boundaries. For some of the edges, there may be WS-Agreement Initiators and Providers at the opposite ends of the edge.

In Figure (3), the edge labeled 1 is meant to capture the following concept: there mechanisms for the application (or the Grid infrastructure in its behalf, e.g. a broker) to invoke services, and pass on to these services parameters like data rate profile (time vs. rate), total data amount remaining (estimation or actual), and other characteristics associated with the data stream and/or the service request. In turn, these parameters aid network services in predicting and optimizing the utilization of network resources, resulting in greater satisfaction and/or cost efficiencies to the end user.

With regard to the edge labeled 2, a designated service must notify an application (or the Grid infrastructure) of those events that the application has negotiated and registered for. It must tell an application if it is admission-controlled out (be it a capacity or a policy issue). It must provide timely notifications of SLA violations to an application .

With regard to the edge labeled 3, when appropriate, credited services can dynamically (re)provision network aspects (e.g., to tap on either traffic engineering fixtures TDM/WDM circuits upon a very large bulk)

The Grid Network Services are the boxes and their derivatives which are directly exposed

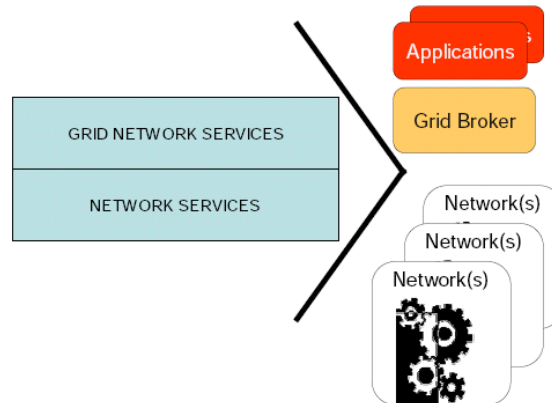


Figure 4: Grid Network Services vs. Network Services

to elements of the Grid infrastructure (such as a universal Grid broker for all resource types). For these elements to interoperate, a GGF-sanctioned Grid interface is necessary. On the contrary, the network services are represented by those boxes and their derivatives which are not directly exposed to elements of the Grid infrastructure (e.g., they only interact with other network services, network control planes management planes). For these, a GGF-sanctioned interface is a sufficient albeit not necessary choice. Examples of network services include: a domain-specific bandwidth broker, a network directory service. It is appropriate to think of the network services forming a practical underlay to the actual Grid Network Services, as pictured in Figure (4).

2.7.2 Network Monitoring Service (a proxy to NM-WG)

Traditionally, network monitoring has been driven by the need for fault detection and performance prediction. While this remains true in Grid environments, a significant new concept is introduced, that of publishing performance data to Grid applications, middleware and the network fabric. This radical change will allow systems to both adapt to changing network conditions, thus optimising performance, and also provide support for the Grids much touted self-healing capability. As figure (5) shows, the services potential clients are numerous varied: middleware and end-user software (Grid applications), other services (e. network cost function), network administration software, such as tools used by human administrators in network operation centre environments, automated systems, and finally, corresponding monitoring services other network domains.

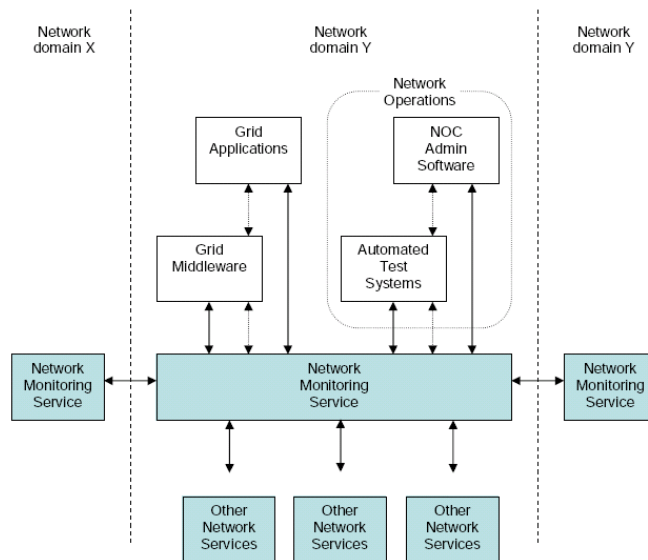


Figure 5: Clients of the Network Monitor Service

2.8 Requirements of a Grid Network service

In their draft [7] Travostino et al. sketch some outlines about what characteristics a Grid Network Service should provide. In the most general case a Grid Network service is supposed to provide:

1. historic performance data, from the running of previous tests
2. real-time performance data
3. new measurements, which may lead to the running of tests
4. future performance data, based on the assumption that a test is already scheduled
5. future performance data, as a prediction
6. event notifications, as a similar concept to SNMP traps

Even at this high-level view of requirements, there are already several points of note:

- In relation to points 1 and 4 above, regularly scheduled tests will need to be performed to provide users with data or predictions relating periods where they have not requested the running of tests.

- Points 3-5 above imply that it should be possible for a user to select whether a data request will ever result in the running of a test.
- As figure (5) suggests, it is expected that requests will work across multiple administrative domains. In addition to this direct requirement, it is clear monitoring services will also need the ability to discover further monitoring services.
- Monitor services could be decompose into several sub-services.
- Further, many of the detailed requirements will make reference to services and monitoring points. These services could be the network monitoring service as a whole, or one of its possible sub-services. Services control monitoring points, the entities which make actual performance measurements. Services and monitoring points have one-to-one or one-to-many relationships.

And in terms of making requests and receiving results:

- Requests for data and tests, and the publication of performance data should in the main make use of the work of the GGF NM-WG group [8] who have defined XML schemas for such tasks.
- Internally, a monitoring service can use any communication method deemed appropriate, but the NM-WG approach should be supported externally. An example of internal communication is that between a monitoring service and its monitoring points (the nodes that actually make measurements).
- Interim communication, that taking place between a request being made and a result being returned, is yet to be addressed.
- A means for requesting event notifications is yet to be defined. In the strictest sense, event notification is a monitoring not measurement task, and may be deemed by NMWG to be outside their scope.

2.9 Web service/Grid Service Definition

From the implementation point of view web services are self-contained, self-describing, modular applications that can be published, located, and typically (but not necessarily) invoked using standard HTTP over port 80. Web services can perform functions that are anything from simple requests to complicated business or scientific procedures.

The W3C Web services Architecture working group provides the following definition: A Web service is a software system designed to support interoperable machine-to-machine interaction over a network. It has an interface described in a machine-processable format (specifically WSDL). Other systems interact with the Web service in a manner prescribed by its description using SOAP messages, typically conveyed using HTTP with an XML serialization in conjunction with other Web-related standards. The main difference between a normal remotely-invoked application and a Web service is that the latter has an XML-based interface description that enables it to be self-describing. Once a Web service component is deployed, other applications can discover and invoke the published service via its interface. Web service are a hence a loosely coupled distributed computing technology but seems to be the choice for the future because:

- Web Services are platform-independent and language-independent, since they use standard XML languages. This means that the client program can be programmed in C++ and running under Windows, while the Web Service is programmed in Java and running under Linux.
- Most Web Services use HTTP for transmitting messages (such as the service request and response). This is a major advantage if you want to build an Internet-scale application, since most of the Internet's proxies and firewalls won't mess with HTTP traffic (unlike CORBA, which usually has trouble with firewalls)

A Grid service is a WSDL-defined service that conforms to a set of conventions relating to its interface definitions and behaviors. OGSA specifies three conditions for a Web service to be qualified as a Grid service. First it must be an instance of a service implementation of some service type as described above. Second, it must have a Grid Services Handle (GSH), which is a type of Grid URI for the service instance. The GSH is not a direct link to the service instance, but rather it is bound to a Grid Service Reference (GSR). The idea is that the handle provides a constant way to locate the current GSR for the service instance, because the GSR may change if the service instance changes or is upgraded. Third, each Grid Service instance must implement a port called GridService portType. This portType is analogous to the base Object class within object-oriented programming languages such as Smalltalk or Java in that it encapsulates the root behavior of the component model. The behavior encapsulated by the GridService portType is that of querying and updating against the serviceData set of the Grid service instance, and managing the termination of the instance.

2.10 Related work

Many researchers have found that high-performance Grid applications need to be aware of the interconnection network in use [17],[18],[19],[20] Performance monitoring systems thus are becoming integral parts of Grid computing platforms.

2.10.1 Network weather service

The Network Weather Service [16] is a distributed system that periodically monitors and dynamically forecasts the performance that various network and computational resources can deliver over a given time interval. The service operates a distributed set of performance sensors (network monitors, CPU monitors, etc.) from which it gathers readings of the instantaneous conditions. It then uses numerical models to generate forecasts of what the conditions will be for a given time frame.

The NWS has been developed for use by dynamic schedulers and to provide statistical Quality-of-Service readings in a networked computational environment. Prototype implementations for Globus and the Global Grid Forum (GGF) Grid Information System (GIS) architecture have been developed.

Currently, the system includes sensors for end-to-end TCP/IP performance (bandwidth and latency), available CPU percentage, and available non-paged memory. The sensor interface, however, allows new internal sensors to be configured into the system.

NWS support various types of forecasting methods: *mean based*, *median based*, *autoregressive method*

The system tracks the accuracy (using prediction error as an accuracy measure) of all predictors, and uses the one exhibiting the lowest cumulative error measure at any given moment to generate a forecast. In this way, the NWS automatically identifies the best forecasting technique for any given resource. Moreover, as new methods are added, they will automatically be used to forecast the resource performance for which they are the most accurate.

2.10.2 NWS: drawbacks

Unfortunately, the information provided by NWS is insufficient for applications with communication patterns where multiple sites compete for the same links. For example, if two pairs of communicating sites are used, NWS will predict performance for each pair separately. If the two communication streams have some links in common, however, these predictions will clearly be too optimistic, as the bandwidth for these links has to be shared

by the two streams. The current NWS system cannot solve this problem, because it lacks topology information and thus cannot know which links are shared. This problem is especially important for applications that use collective communication, where many sites communicate with each other simultaneously. This problem has been already faced in section 2.2.5 where the approach of Lancur et Al. has been explained. Even though their approach could make sense and make feasibly the description of large networks otherwise infeasible, for small networks it could lead to less accuracy in description w.r.t the approach taken by Topomon and by the present work. The shared links problem will be investigated in details in next sections.

2.10.3 Topomon

TopoMon [6] is a tool which heavily inspired the present work. It was made to augment NWS with topology information about the wide-area network. It is based on the idea that the added value of integrating both performance and topology information is that applications or communication libraries can not only predict their communication performance, but can also avoid congestion on those Internet links that are shared by multiple, simultaneous data streams.

2.10.4 Other kinds of approach

There exist several monitoring systems for the Internet that either measure network performance [21],[23], [24] or explore network topology [25], [26]. However, their usefulness for Grid applications is limited as they only provide data for the nodes involved in their own monitoring efforts. In order to have a useful information for Grid applications some information (topology and performance) about exactly those parts of the global Internet that connect the sites of a Grid environment is needed. While Internet-related tools provide useful monitoring techniques, Grid environments have to apply them to their own sites. Some systems explore the topology of LAN installations in great detail. Most prominently, Remos [17] uses SNMP to collect the necessary information. However, SNMP is not available for general Internet connections. The BGP [22] routing protocol provides global topology information, however only with the granularity of Autonomous Systems, which is insufficient for our purposes.

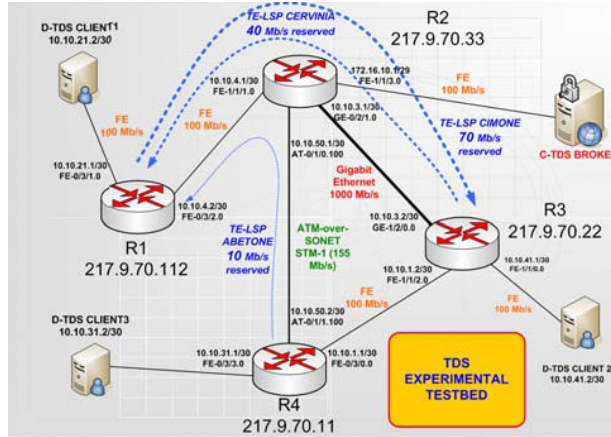


Figure 6: Testbed for D-TDS

3 The proposed approach

This section describes the architecture and the implementation issues of the system for topology discovery presented. The topology discovery service, following the recommendation proposed in 2.8, provides grid users with up to date information on the grid network status considering different network layers. The proposed implementation is based on a producer/consumer architecture developed under the GT3 framework Here onwards the system taken into consideration will be referred as D-TDS (Distributed topology discovery service).

3.1 Architecture

The Distributed TDS (D-TDS) architecture, as depicted in Figure (6), consists of a set of grid hosts (i.e., users) and a broker. The grid hosts are both producers and consumers of the information provided by the topology discovery service. The hosts run a service capable of discovering the link-by-link paths to other hosts belonging to the Virtual Organization (VO). The D-TDS provides the grid users with the logical topology, i.e. with the router adjacencies at the IP layer. IP layer adjacencies consist of both physical layer adjacencies (i.e., the adjacencies between physically connected routers) and MPLS layer adjacencies (i.e., two routers non physically adjacent can be adjacent at the IP layer because connected by a Label Switched Path (LSP)). Logical topology is considered as something between the physical and functional topology treated in 2.2.4.

In the D-TDS architecture the D-TDS broker tasks are just to trigger a sub-set of network hosts to produce the requested information and to gather and elaborate the information

once it is produced. Thus, the D-TDS broker does not implement any communication with the routers but it communicates just with the hosts. In addition the D-TDS broker does not require to have any administrative privileges on the routers. The topology discovery service procedure is as follows.

1. A user (i.e., hosts) submits a query to the broker regarding the VO logical network topology. More precisely it ask the broker to drive measurement between a pool of hosts whose IP is given as input to the broker
2. The broker contacts each hosts asked by the client triggering the topology discovery procedure.
3. Each contacted host runs the sensor command to find the path, on the logical topology, to any other host in the set.
4. Upon termination of the sensor-based discovery process, each host sends to the broker the tree rooted on itself to every other host in the selected host set.

In particular, if a shortest path protocol, such as OSPF, is utilized to route IP packets between the VO routers, the tree provided by each host to the D-TDS broker is the shortest path tree (in terms of router hops) from itself to all the other hosts in the set. Upon reception of all the host-rooted trees the broker merges them to compute the logical network topology. Eventually the broker sends to the host that triggered the topology discovery service the reply containing the computed VO logical network topology.

3.2 Implementation

In the implementation of the proposed D-DTS, the D-DTS broker runs on one of the VO host. This does not imply any loss of generality since the broker and the topology discovery service can run independently. The services developed for implementing the D-DTS topology discovery service are grid services implemented in GT3.2 and deployed in the globus container web server. All the communication between D-DTS broker and hosts and between hosts is implemented via web service SOAP through the `Globus Toolkit` libraries. Both the D-DTS broker and the topology discovery service at the hosts are implemented in `java(tm)`. The topology discovery service has only one remote method `getTree` which is called by the broker and takes a host list as input (i.e., the set of selected hosts involved in the distributed topology discovery process) which turns to be a simple list of host IPs. The method `getTree` returns the list of the links discovered through `pathchar` attributed with bandwidth, loss percentage, round trip time and queue delay per link.

As it was mentioned before `pathchar` offers a way of estimating the available bandwidth in a non invasive fashion differently from other tools, such as `iperf` and `netperf`, that introduce a huge amount of traffic to give an estimate of the link bandwidth. Moreover as it was mentioned before, `pathchar` is able to measure the network link characteristics on a per-link basis. Instead a lot of tools (such as `bing` or `cprobe`, `bprobe`) give an aggregate end-to-end information. `Pathchar` is run by the grid service on each host once the method `getTree` is called and the list of the nodes in the pool is provided by the D-DTS broker. It is important to notice that the method is called by the D-DTS broker asynchronously with respect to each topology discovery service on the hosts of the pool so that each host can start the measurements almost contemporarily, thus speeding up the overall process. This asynchronous procedure call has been implemented in java via the `Thread` package. Once the remote method is called, each host starts collecting measurements. Each host involved in the measurements process then parses the result returned by `pathchar` and extracts a formatted information to be returned to the broker. The parser is written utilizing the java `StreamTokenizer` package. The hosts return the information to the broker via grid notification. In fact each hosts fills up a predetermined SDE (service data element) and sends it to the broker which subscribed to it before calling the `getTree` method. Once the broker has gathered all the information from all the selected hosts, it can process it and sends it to the host that requested the service by means of a `XML` file.

A more detailed description of the implementation issues will be provided in the appendix [B](#)

3.3 GT3

The D-TDS has been implemented as a pair producer/consumer where the topology discovery service has been developed under GT3 (Globus Toolkit) This section aims to explain to some extent the taxonomy and the feature provided by Globus in relation to the service here presented

3.4 About the Globus Toolkit

The open source Globus Toolkit [\[28\]](#) includes software services and libraries for resource monitoring, discovery, and management, plus security and file management.

The toolkit includes software for security, information infrastructure, resource management, data management, communication, fault detection, and portability. It is packaged as a set of components that can be used either independently or together to develop applications.

Every organization has unique modes of operation, and collaboration between multiple organizations is hindered by incompatibility of resources such as data archives, computers, and networks. The Globus Toolkit was conceived to remove obstacles that prevent seamless collaboration. As it was mentioned before, its core services, interfaces and protocols, allow users to access remote resources as if they were located within their own machine room while simultaneously preserving local control over who can use resources and when. More information on the Globus Toolkit can be found here [2] [3]

3.5 Globus installation, testbed setup

In order to test the D-TSD system a grid-enabled testbed has been set up.

The testbed consist of 3 hosts connected through different routers as depicted in (6) The hosts run a standard Linux distribution (Debian and RH9) and `Globus Toolkit 3.2` has been installed on them. The installation process has not been too much plain as it will be explained later. The new release of Globus (GT4) improves to a great extent this issue. In appendix A are reported some tricks to speed up thee installation process. This howto doesn't mean to be an installation tutorial, it is just a memo made of (hopefully) useful information for grid newbies.

3.6 Further consideration about the discovered topology

In the topology information collected by the broker it possible (highly probably, indeed) that a link is reported more than once to the D-TDS broker by different hosts involved in the topology discovery procedure because the same link can be shared by many paths between different host pairs. This information, regarding the “collision” of different path is indeed very important as it was mentioned since provides more knowledge about the dynamic setting of the grid network, that is, it considers that competitive traffic could be present. An intuitive way of elaborating the redundant link parameter information from the quantitative point of view is to consider the duplicated links as a single link which connects the same pair of nodes. Thus the quantitative performance measurements, such as bandwidth and delay, can be obtained, for example, by averaging the set of values reported for the duplicated links between the same node pair.

It is important to underline that the topology discovery service based on the D-TDS architecture presents the drawback of not always guaranteeing the discovery of the complete VO network. Indeed the set of users among which the trees are discovered might not coincide with the minimum set of hosts necessary for the complete network topology discovery

or, even though their number could be sufficient, their placement cannot guarantee the complete network topology discovery. For example in (6), if the hosts producing the information are connected to the routers R1, R2, and R3, all the links in the networks are not discovered by the D-TDS. (Specifically the link R4-R2 is not discovered.) Instead if the hosts are connected to the routers R1, R3, and R4 all the network links are discovered. On the other hand the distributed approach adopted in the D-TDS represents also an advantage. Indeed the possibility of selecting the number of hosts involved in the topology discovery process allows to trade-off the level of details of the discovered topology with the amount of the resources (i.e., messages) utilized for the discovery. This makes the D-TDS scalable to MAN with an elevated number of network elements. Fourth, it can be claimed that for discovering the complete logical network topology it would be sufficient that at least one host of the VO is connected to each VO network element and that each host just notifies to the broker its neighbors. However this scenario is unlikely because, in general, VOs span various autonomous systems and in a specific VO some routers can be just transit routers without any connected host.

On the other hand the Distributed-TDS (D-TDS) allows to obtain information on the topology traversed by IP packets, i.e. the logical topology, that depends on the number of users (i.e., grid hosts) involved in the discovery procedure. However the D-TDS is scalable and it utilizes tools available to regular grid users, such as `pathchar` .

4 Results

For what it concerns the D-TDS implementation, below is shown the output of the `pathchar` command run from host 10.10.21.1 (CLIENT1) to host 10.10.31.2 (CLIENT3). It can be noticed that `pathchar` announces just the input interface of the crossed network element.

```
pathchar to 10.10.31.2 (10.10.31.2)
doing 3 probes at each of 45 sizes (64 to 1500 by 32)
0 localhost
| 54 Mb/s, 143 us (508 us)
1 10.10.21.1 (10.10.21.1)
| 95 Mb/s, 16 us (668 us)
2 10.10.4.1 (10.10.4.1)
| ?? b/s, 233 us (1.13 ms)
3 10.10.50.1 (10.10.50.1)
| 87 Mb/s, -34 us (1.20 ms), 15% dropped
```

4 10.10.31.2 (10.10.31.2)

4 hops, rtt 716 us (1.20 ms), bottleneck 54 Mb/s, pipe 9848 bytes

An excerpt of the D-TDS logical network topology is depicted below. The formatted XML file is obtained by providing the broker with the information regarding which interfaces (i.e., IP addresses) belong to which routers and by assuming that all the interfaces of the router are fully connected within the router. In the XML file are present some tag `<nodes>` which represent the IP of the selected hosts. After the node stanza a list of links is provided: these are all the links, comprehensive of the measured bandwidth, round trip time, queue delay and dropped packets computed by the services through pathchar. The D-TDS based topology discovery service is able to obtain an estimate of the link bandwidth and of the round trip time (on which the latency depends). However the D-TDS is not able to obtain information on each router interface as it was mentioned before. In addition, the link bandwidth appears underestimated, especially for high capacity links. This can be due to the pathchar specific implementation.

```
<D-TDS-topology>
<node>
<name>10.10.21.2</name>
<dns-name>ringo</dns-name>
<node-type>host</node-type>
<link>
<source>10.10.21.2</source>
<destination>10.10.21.1</destination>
<dns-destination>R1</dns-destination>
<timestamp>12:34</timestamp>
<available-bandwidth>50 Mb/s</available-bandwidth>
<rtt>7.1</rtt>
<dropped>5</dropped>
<queue-delay>1.3E-4</queue-delay>
</link>
</node>
<node>
<name>217.9.70.112</name>
<dns-name>R1</dns-name>
<node-type>router</node-type>
<link>
```

```

<source>10.10.21.1</source>
<destination>10.10.21.2</destination>
<dns-destination>ringo</dns-destination>
<timestamp>12:34</timestamp>
<available-bandwidth>50 Mb/s </available-bandwidth>
<rtt>7.1</rtt>
<dropped>5</dropped>
<queue-delay>1.3E-4</queue-delay>
</link>
<link>
<source>10.10.4.2</source>
6
<destination>10.10.4.1</destination>
<dns-destination>R2</dns-destination>
<timestamp>12:33</timestamp>
<available-bandwidth>87 Mb/s</available-bandwidth>
<rtt>7.43</rtt>
<dropped>0</dropped>
<queue-delay>1.6E-4</queue-delay>
</link>
</node>

```

4.1 Considerations about running time

An important characteristic that has been neglected so far is about the time required by the system to perform a complete round of measurements given a pool of hosts.

This is a very important issue for two main reasons:

1. since the measurements have to be performed periodically in order to keep the grid network description up to date, the time it takes the system to perform them limits the maximum frequency at which measurements could be taken.
2. in the proposed infrastructure an host submits a request for measurements among a pool of hosts. Each host in the pool have to perform measures towards every other host in the pool and measures are performed on a per-hop basis and in parallel. The total time needed by the $i - th$ host for a measurement is then (neglecting the time

unit of measurement):

$$T(i) = \sum_{j=1, i \neq j}^{j=H} l_{i-j} \quad (1)$$

where l_{i-j} is the length of the path (numbers of hop) from i to j (assuming that the time needed to test a hop to be a constant) and H is the number of host towards which the measure be performed.

Since each host computes the measurement in parallel then the total time T can be approximated as $\max(T(i))$

The 1 can be rewritten as:

$$T \leq (H - 1) * \max(E[l_{i-j}, \forall j]) \forall i \quad (2)$$

where $E[l_{i-j}, \forall j]$ is the average path length for each pair of host $i - j$ taken w.r.t j in the pool and the operator \max is computed on the index i

It would be interesting to give some asymptotic bounds valid on very large networks. It has been noticed [30] that network topology, taken both at router level and at autonomous system level usually exhibit the so called “small-world” property. It is also known [31] that if a graph exhibits small world property then its diameter D (defined as the longest among the shortest paths between all node pairs) is proportional to the logarithm of the number of nodes in the graph. Provided that, 2 becomes (the Big-O notation is used):

$$T \leq (H - 1) * D = O(H \log(N)) \quad (3)$$

Where N is the number of hosts in the network considered and H is still the number of hosts for which the measurements are performed.

It comes out that the time needed for a complete round of measurement is then proportional to the number of host involved in the measurement in a particular network and increases linearly w.r.t logarithm of the total number of hosts in the network.

Testing the tool on the network depicted in (6) it emerges that it takes 4 minutes for a complete round of measures.

The time needed depend indeed also on the precision at which `pathchar` probes the paths to and fro host. A greater accuracy in measurement would require more time.

In order to speed up the process other network sensors instead of `pathchar` could be taken into consideration, in any case, since the asymptotic bound given before, the system doesn't scale to large network (where N as well as H are high). In this case, in order not to take too much time to perform the measurements, H should be reduced w.r.t. to N thus reducing the completeness of the topology discovered since it increases as the ratio $\frac{H}{N}$ increases, ideally tending to 1. Anyway, as it was mentioned before, a witty choice of the hosts to include in the pool submitted for measurements could help a lot, for instance, including the host on the border of the network could be a good choice.

4.2 Future work

The proposed system could be improved to a great extent both from inside (adding some features) and outside (adding new intermediate piece of software to elaborate its results) point of view.

4.2.1 Security

At first the system should be made secure. The communication between hosts is now in plain text with no authentication.

Security tools are concerned with establishing the identity of users or services (authentication), protecting communications, and determining who is allowed to perform what actions (authorization), as well as with supporting functions such as managing user credentials and maintaining group membership information.

The forthcoming version of the `Globus Toolkit` (GT4) provides distinct WS and pre-WS authentication and authorization capabilities. Both build on the same base, namely standard X.509 end entity certificates and proxy certificates, which are used to identify persistent entities such as users and servers and to support the temporary delegation of privileges to other entities, respectively.

More information about security in grid environment can be found here: [\[32\]](#)

4.2.2 Network sensor improvements

A considerable effort should be spend in finding a better network sensor w.r.t. `pathchar`. As it was mentioned before, `pathchar` features many advantages but as a drawback it cannot handle high capacity links with accuracy and, since links throughput is constantly

increasing in the time, `pathchar` cannot be the solution for the future.

New network sensor have thus to be extensively tested in order to find a better solution.

4.2.3 Grid topology service

Even though the server-side of the proposed infrastructure, except for the modification proposed before, can be considered somehow self-contained and complete, the broker can be improved to a great extent.

In this work the broker has been considered as joint with the client itself but its final purpose should be that to provide a persistent middleware able to answer to queries submitted by grid users about topology.

In order to do this the broker should become a standalone grid-service able to query a pool of hosts that communicate to each other in a peer-to-peer fashion to compute the relative topology.

The broker could then be empowered by a historic data server, able to store past measures and topology and forecast those quantities for the future like the Network Weather Service but empowered with topology awareness.

A Globus installation tricks

Installing and configuring GT3.2 could seem a tricky task to be accomplished. A GT3.2 installation and administration guide can be found here [33]. At the first chapter of this guide a plenty of support software is required to be installed before GT3.2 can be installed as well.

The aim of this section is to sketch a list of best practices for GT3.2 installer newbies.

- If it is possible installation from source should always be chosen . Binary package is provided as well but it is highly likely that it won't work perfectly. GT3.2 is a huge piece of software, in part written in `java`, `perl` and `bash` and thus decoupled from OS but in part (i.e. the GridFTP bundle) very architecture dependent. Building from source requires some additional software (i.e. C compiler) but it is guaranteed to work properly.
- The instructions provided have to be followed carefully (this is a rule of thumb for doing whatever, indeed), because sometimes globus works differently as expected. For instance, in order to start the globus container the command `globus-start-container` have to be issued from the globus root directory, otherwise it won't work properly. This is really counter-intuitive.
- If something goes wrong, always googling the error reported will point to some globus-discuss forum where hopefully a solution for the problem will be given. The fora about globus are very active on the Internet and sometimes could help more than manuals and tutorial.

B Programming GT3

Writing a Grid Service for GT3 is an exercise that requires some background notions. A great tutorial by Borja Sotomajor can be retrieved here [29]

In this section a coarse grain introduction to programming GT3.2 and grid services is provided.

According to [29] writing and deploying a Grid Service could be done following five simple steps.

1. Define the service's interface. This is done with GWSDL
2. Implement the service. This is done with `java`

3. Define the deployment parameters. This is done with WSDD
4. Compile everything and generate GAR file. This is done with `ant`.
5. Deploy service. This is also done with `ant`.

B.0.4 OGSA,OGSI,GT3 and WSRF

Before going on explaining how to program a grid service some basic concepts have to be drawn.

The Open Grid Services Architecture (OGSA), developed by The Global Grid Forum, aims to define a common, standard, and open architecture for grid-based applications. The goal of OGSA is to standardize practically all the services one finds in a grid application (job management services, resource management services, security services, etc.) by specifying a set of standard interfaces for these services.

OGSA alone doesn't go into much detail when describing Grid Services. It basically outlines what a Grid Service should have (that Web Services don't) but little else. That is why OGSA spawned another standard called the Open Grid Services Infrastructure (OGSI, also developed by The Global Grid Forum) which gives a formal and technical specification of what a Grid Service is.

The Globus Toolkit as it was mentioned in section 3.4 is a software toolkit, developed by The Globus Alliance, which can be used to program grid-based applications. The third version of the toolkit (GT3) includes a complete implementation of OGSI. However, it's very important to understand that GT3 isn't only an OGSI implementation. It includes a whole lot of other services, programs, utilities, etc. Some of them are built on top of OGSI and are called the WS (Web Services) components, while other are not built on top of OGSI and are called the pre-WS components.

OGSI specification has shown to have different drawbacks so to solve OGSI's problems and to improve Grid Services' chances of finally converging with Web Services, a new standard was presented in January 2004 to substitute OGSI: The Web Services Resource Framework, or WSRF.

WSRF aims to integrate itself into the family of Web Services standards, instead of simply being a 'patch over Web Services' like OGSI was. In this new scenario, OGSA will be based directly on Web Services instead of being based on OGSI Grid Services.

B.0.5 The five steps

- The first step in writing a Grid Service (or a Web Service) is to define the service interface. It is necessary to specify what the service is going to provide to the outer world. This description is not concerned with the inner workings of that service (what algorithms it uses, what databases it will access, etc.) In Web/Grid Services lingo, the service interface is usually called the port type (usually written `portType`).
- After defining the service interface ("what the service does"), the next step is implementing that interface. The implementation is "how the service does what it says it does". The implementation of a Grid Service is simply a `java` class which, besides implementing the operations described in the GWSDL file, has to meet certain requirements. This class can furthermore include additional private methods which won't be available through the Grid Service interface, but which the service can use internally.
- Up to this point, the two most important parts of a Grid Service have been written: the service interface (GWSDL) and the service implementation (`java`). Now this pieces have to be put together and made available through a Grid Services-enabled web server. This step is called the deployment of the Grid Service.
- One of the key components of the deployment phase is a file called the deployment descriptor. It's the file that tells the web server how it should publish the Grid Service (for example, telling it what the the service's GSH will be). The deployment descriptor is written in WSDD format (Web Service Deployment Descriptor).
- in this step everything comes together in perfect harmony. The three files created before have to be put together in a Grid Archive, or GAR file. This GAR file is a single file which contains all the files and information the grid services container need to deploy the service and make it available to the whole world.

The GAR file contains all the files and information the web server needs to deploy the Grid Service. Deployment is done with the Ant tool, which unpacks the GAR file and copies the files within (WSDL, compiled stubs, compiled implementation, WSDD) into key locations in the GT3 directory tree. also reads our deployment descriptor and configures the web server to take our new Grid Service into account.

B.1 Deeper inside implementation

As mentioned before, each host runs a topology discovery service and has the `pathchar` tool installed with proper privileges set.

B.1.1 Server-side

The service has only a public method: `getTree`

```
public void getTree(NetDiscoveryServerIPArrayType serversIP)
```

This method takes as input the list of the hosts's IP addresses that have to be tested to each other. The method doesn't return anything since the results are given back to the broker via notification.

The datatype `serversIP` is defined in an XML schema like this:

```
<complexType name="NetDiscoveryServerIPArrayType">
    <sequence>
        <element name="serverIP" type="string" minOccurs="1" //
            max Occurs="unbounded"/>
    </sequence>
</complexType>
```

The topology discovery service runs the `pathchar` tool for each IP in the `IPList` provided by the broker then grabs and parses its output. With the values parsed the service fills a service data and gives it back to the client via notification.

The SDE filled and returned to the broker has the following structure:

```
<complexType name="LinkStatusDataType">
    <sequence>
        <element name="IP_a" type="string"/>
        <element name="IP_b" type="string"/>
        <element name="timestamp" type="string"/>
        <element name="maxthroughput" type="float"/>
        <element name="linkbw" type="float"/>
        <element name="capacity" type="float"/>
        <element name="rtt" type="float"/>
        <element name="dropped" type="float"/>
    </sequence>
</complexType>
```

```

                <element name="queue_delay" type="float"/>
            </sequence>
</complexType>

```

IP_a and IP_b are the source and destination IP of the link measured. timestamp is the time at which the measurements has been taken
pathchar provides data to fill up only the fields linkbw, rtt, dropped and queue_delay which represents the link bandwidth, round trip time, percentage of dropped packets and delay accumulated by packets in router's queue.

B.1.2 Client-side: the broker and the grid end user

The broker program, as it was mentioned before has been joint to the user program for convenience.

The broker, once received a request for measurements from the user (in the proposed infrastructure, since the broker and the user are the same think, the broker simply reads input data from the command line) triggers the method `getTree` for all the IP hosts provided by the user.

The remote procedure call of the `getTree` method on different hosts is done in parallel and asynchronously via distinct java Threads.

```

for( int i = 0; i < serverList.size(); i++ ) {
    HandleType ht= (HandleType)handleList.get(i);
    sink = notifManager.addListener( "NetDiscoveryData", null,
        (HandleType)handleList.get(i) , this );
    sinkList.add(sink);
    GetTreeThread gt=new GetTreeThread(netDiscoveryLocator.
        getNetDiscoveryServicePort( (HandleType)handleList.get(i)
        ),serverList,i);
    gt.start();
    threadList.add(gt);
}

```

Before calling the remote procedure for each host in the IPlist a listener for the notification is activated.

Once notifications have come back to the broker from the hosts contacted, the broker, acting as a client, delivers on standard output the informations gathered encapsuled in XML

```

public synchronized void deliverNotification( ExtensibilityType any )
    throws RemoteException {
try {

    ServiceDataValuesType serviceData = AnyHelper.getAsServiceDataValues( any );
    LinkStatusDataTypeArrayType monResults = (LinkStatusDataTypeArrayType) //
        AnyHelper.getAsSingleObject(serviceData, LinkStatusDataTypeArrayType.class);

    ...
for( int i = 0; i < monResults.getResultsData().length; i++ ) {
    bw=monResults.getResultsData( i ).getLinkbw();
    ...
    qd=monResults.getResultsData( i ).getQueue_delay();
    ...
    System.out.println("\t<link source=\"" + monResults.getResultsData( i ). //
        getIP_a() + "\" destination=\"" + monResults.getResultsData( i ). //
        getIP_b() + "\" timestamp=\"" + monResults.getResultsData( i ). //
        getTimestamp() + "\" available_bw=\"" + bws + "\" rtt= \"" + //
        monResults.getResultsData( i ).getRtt() + "\" dropped=\"" + //
        monResults.getResultsData( i ).getDropped() + "\" //
        queuedelay=\"" + qds + "\"/>");

    ...

    }
} catch( Exception e ) {
    System.err.println("Errore: " + e.getMessage());
    e.printStackTrace();
}
}
}
}

```

In the previous excerpt some parts are omitted for the sake of simplicity but it is still possible to reckon the structure. The function `deliverNotification` is automatically called

once a service data filled by some topology service has been returned to the client.

The function has been declared `synchronized` since the topology discovery services could return data concurrently thus calling the function in the same time and thus leaving internal data inconsistent.

At first the function casts the variable passed to the function as an object of type `LinkStatusDataTypeArrayType` and then in the body of the cycle for the service data returned is parsed and printed in XML

C Grid Topology service installation and user guide

This section aims to provide a comprehensive list of the operation to be performed to compile, deploy, run and access the topology discovery service.

The instruction here provided are meant to work under UNIX-like environment, however, since both the client and the service are implemented in java which is a platform-independent language, it should not be hard to have them up and running on other OS as well.

C.1 Anatomy of the directory

Source code and `gar` package of the system are provided in a `tar.gz` archive: `NetDiscovery.tar.gz`

Once decompressed, the archive has appears as follow:

```
./NetDiscovery
./NetDiscovery/NDservice
./NetDiscovery/NDservice/core
./NetDiscovery/NDservice/core/first
./NetDiscovery/NDservice/core/first/NetDiscoveryService
./NetDiscovery/NDservice/core/first/NetDiscoveryService/client
./NetDiscovery/NDservice/core/first/NetDiscoveryService/impl
./NetDiscovery/NDservice/core/first/NetDiscoveryService/schema
./NetDiscovery/example_topology
./NetDiscovery/lib
```

The directory `/NetDiscovery/NDservice/core/first/NetDiscoveryService/` contains 3 subdirectory:

1. `client` contains the client implementation `NetDiscoveryClient.java` and the implementation of the class `GetTreeThread` that handles with threads
2. `impl` contains the service implementation in java `NetDiscoveryImpl.java`
3. `schema` contains the service description `NetDiscovery.gwsdl` and the definition of the datatype needed by the service `NetDiscoveryDataType.xsd`

The deployment descriptor is contained in:

```
./NetDiscovery/NDservice/core/first/NetDiscoveryService/NetDiscovery.wsdd
```

The directory `./NetDiscovery/build` is built and filled by `Ant` in the compilation process. Once the compilation is terminated, the `gar` file to be deployed can be found in the directory: `./NetDiscovery/build/lib/`

C.2 Compiling, deploying and running the service

In order to compile the package, a script `NetDiscovery_build.sh` is provided. Some environment variable have to be set but if no changes are made to the directory tree provided the script should work properly as is.

The steps to compile are:

1. source the `set_classpath.sh` script in order to set the right classpath variables
2. run the `NetDiscovery_build.sh` script, eventually preceded by the command `ant clean` to purge the previously built processes.

Once compiled the service, if no errors are reported it is possible to deploy it.

Steps to deploy the `gar` file are:

1. log as the `globus` user on the machine you want to deploy the service.
2. `cd` into the `globus` root directory (e.g. `/usr/globus/`)
3. issue the command:

```
ant deploy -Dgar.name=$NDservice_root/build/lib/
```

`NDservice.core.first.NetDiscoveryService.NetDiscovery.gar` where the variable `$NDservice_root` has to be replaced with the full path where the `.tar.gz` package was extracted. This example take into consideration the scenario where the `gar` file and `globus` are on the same machine.

4. The service could be undeployed just issuing the command:

```
ant undeploy -Dgar.id=NDservice
```

In order to start the `globus` container with the new service built-in it is sufficient to:

1. log as the `globus` user on the machine you want to deploy the service.
2. cd into the `globus` root directory (e.g. `/usr/globus/`)
3. issue the command: `bin/globus-start-container`

It is important to notice that the command have to be issued from the root directory otherwise it won't work properly.

In order to have the service running properly `pathchar` should be installed in the `/usr/local/bin` directory on each machine which runs the service. `pathchar` must be `suid` root (`chmod u+s /usr/local/bin/pathchar`)

C.3 Compiling and running the client

The broker-client can be compiled just using the java compiler `javac`:

1. source the `set_classpath.sh` script in order to set the right classpath variables
2. issue the command:

```
javac -classpath .:build/classes/:$LOCALCLASSPATH
```

```
NDservice/core/first/NetDiscoveryService/client/NetDiscoveryClient.java
```

The broker-client can be run from the command line providing the list of the hosts' IP to test on standard input. An example of a possible command can be:

```
cat Iplist.txt | java -classpath .:build/classes/:$LOCALCLASSPATH
```

```
-Dorg.globus.ogsa.schema.root=http://localhost:8080/
```

```
NDservice/core/first/NetDiscoveryService/client/NetDiscoveryClient
```

where the file `Iplist.txt` is supposed to contain the list of the hosts's IP to be tested one per line.

The broker-client returns the measures performed to standard output in XML format and some debug information to standard error.

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